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# **Unslotted Hybrid CDMA/ISMA Protocol for Indoor Wireless Computer Communications**

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### Abstract

A hybrid Code Division Multiple Access / Inhibit Sense Multiple Access (CDMA/ISMA) protocol has been proposed as an effective multiple access scheme for Indoor Wireless Computer Communications. This new protocol combines the advantages of both CDMA and ISMA into one protocol. On the one hand the ISMA protocol introduces a limitation to the number of simultaneous accesses to the transmission channel. On the other hand the CDMA protocol introduces an improvement to the packet survival chance.

It is shown that the performance of the hybrid protocol is indeed better than CDMA only. In addition, code sharing can be applied to reduce hardware cost. The slotted hybrid CDMA/ISMA protocol using the *p-persistent* ISMA scheme has already been analysed. In this report, we investigate the hybrid protocol using the *unslotted non-persistent* ISMA scheme. The analysis is done for a star-connected multiple access wireless computer network. The performance comparison between the slotted and unslotted hybrid CDMA/ISMA protocol is evaluated in terms of throughput and delay using computer simulation and mathematical analysis.

Indexing Terms: Code Division Multiple Access (CDMA), Inhibit Sense Multiple Access (ISMA), Spread Spectrum, Data Communications, Rayleigh Fading.



## SUMMARY

Compared to the fixed wired systems, Indoor Wireless Communication offers advantages such as time and cost saving, reduction or elimination of wiring and rewiring and increased flexibility and efficiency. However, the scarce electromagnetic spectrum and the hostile transmission medium oblige us to choose an appropriate multiple access protocol for the indoor wireless communication.

Herefore, a hybrid Code Division Multiple Access (CDMA) / Inhibit Sense Multiple Access (ISMA) protocol has been proposed. This protocol combines the advantages of both CDMA and ISMA into one protocol. The ISMA protocol limits the number of simultaneous transmissions of data packets while CDMA improves the survival chances of the transmitted packets. Furthermore, code sharing can be applied with a reasonable penalty in performance degradation.

We distinguish two types of hybrid protocols: slotted and unslotted. The slotted p-persistent hybrid CDMA/ISMA protocol has already been investigated and reported on. This work concentrates on the unslotted non-persistent hybrid CDMA/ISMA protocol.

The performance is evaluated in terms of throughput and delay in relationship to the offered traffic in a star-connected wireless computer network. The throughput is a measure for the efficiency of the protocol and the delay gives the average delay the data packets suffer before they are correctly received. We adopt two approaches to derive the performance of the hybrid protocol. First, we set up a computer simulation program and next, a mathematical model is derived. Then the results are compared.

As can be expected, comparison between the slotted and unslotted protocol shows that for low offered traffic the delay for the unslotted protocol is better than the slotted one. For high traffic it is the other way round. The hybrid protocol performs very well in case the propagation delay is not a concern. If the propagation delay is too high, the performance drops quickly. The results also show that code sharing is a good option for the hybrid protocol.

# SAMENVATTING

Vergeleken met de kabelsystemen biedt binnenshuis draadloze communicatie voordelen zoals tijd- en kostenbesparing, reductie of eliminatie van bekabeling en herbekabeling en toename van flexibiliteit en efficiëntie. Echter, de schaarste van de elektromagnetische spectrum en de slechte transmissiemedium verplicht ons om een geschikte “multiple access” protocol voor de binnenshuis draadloze communicatie te vinden.

Hiervoor is een hybride Code Division Multiple Access (CDMA)/ Inhibit Sense Multiple Access (ISMA) protocol voorgesteld. Dit protocol combineert de voordelen van zowel CDMA als ISMA in één protocol. Het ISMA protocol beperkt het aantal gelijktijdige transmissies van datapakketten terwijl CDMA de overlevingskansen van de gezonden pakketten verbetert. Bovendien kan hergebruik van codes toegepast worden met een redelijke degradatie in de prestatie.

Wij onderscheiden twee types hybride protocollen: ‘slotted’ en ‘unslotted’. De slotted p-persistent hybride CDMA/ISMA protocol was al onderzocht en gerapporteerd. Dit werk concentreert zich op de non-persistent hybride CDMA/ISMA protocol.

De prestatie is geëvalueerd in termen van doorstroming en vertraging in relatie tot het aangeboden verkeer in een draadloze ster verbindingsnetwerk. De doorstroming is een maat voor de efficiëntie van de protocol en de vertraging geeft de gemiddelde vertraging van de datapakketten voordat zij correct worden ontvangen. We gebruiken twee benaderingswijzen om de prestatie van de hybride protocol te achterhalen. We zetten eerst een computer simulatie programma op en daarna wordt een mathematische model afgeleid. Daarna worden de resultaten met elkaar vergeleken.

Zoals verwacht laat de vergelijking tussen de slotted en de unslotted protocol zien dat voor laag aangeboden verkeer, de vertraging voor de unslotted protocol beter is dan die van de slotted protocol. Voor hoog aangeboden verkeer is het net andersom. De hybride protocol presteert goed in het geval wanneer de propagatievertraging geen rol speelt. Als de propagatievertraging te hoog is, neemt de prestatie snel af. De resultaten laten ook zien dat hergebruik van codes een goede optie is voor hybride protocol.

# PREFACE

This report is the results of my thesis work performed at the Department of Telecommunications and Traffic Control Systems, Faculty of Electrical Engineering, Delft University of Technology.

I would like to thank all my colleague students at the department for the friendly and nice atmosphere and the many joyable discussions during the breaks. Of course, this thesis work would not have been possible without the valuable advices, support and fruitful cooperation I have received from my mentors. This is why I specially like to thank R. Prasad, J.A.M. Nijhof and H.R.R. van Roosmalen.

I'd like to dedicate this report to my parents and Hong Hanh for their encouragements, motivation, support and guidance.

Huy Linh Anh Le

Delft, September 5th 1995

# LIST OF SYMBOLS AND ABBREVIATIONS

$a_k$	Code waveform
$A$	Carrier signal amplitude
$b_k$	The $k$ th user's information signal
$BW_{RF}$	RF bandwidth
$c$	Speed of light
$C$	Capacity in bits per second
$d$	Inhibit delay fraction
$E_b$	Energy per bit
$f$	Signal frequency
$g_T$	Transmitter gain
$g_R$	Receiver gain
$G$	Offered traffic
$G_p$	Processing gain
$K$	Number of users
$K_a$	Number of antennas
$l$	Distance between the transmitter and the receiver
$L$	Number of resolvable paths
$L_p$	Packet length
$M$	Order of diversity
$n(t)$	White Gaussian noise
$N$	Noise power
$M_{max}$	Maximum order of diversity
$P_{tr}$	Probability that a data packet is transmitted in case the channel is sensed free (only for slotted ISMA protocol)
$P_R$	Received power
$P_T$	Transmitted power
$R_{data}$	Data rate

$r(t)$	Received signal
$s_k(t)$	Transmitted signal
$S$	Signal power
$t_c$	Cycle time
$T_b$	Bit duration
$T_c$	Chip duration
$T_{IA}$	inter-arrival time
$T_m$	The rms delay spread
$T_{pd}$	Packet duration
$W$	Bandwidth in hertz
$x$	The number of arrivals during the inhibit delay fraction
$\alpha$	Random delay parameter
$\beta$	Path gain
$\gamma$	Path phase
$\lambda$	Arrival rate
$\tau$	Path delay
$\mu$	Attenuation parameter
$\omega_c$	Carrier frequency
AWGN	Additive White Gaussian Noise
CDF	Cumulative Distribution Function
CDMA	Code Division Multiple Access
CSMA	Carrier Sense Multiple Access
DPSK	Differential Phase Shift Keying
DS	Direct Sequence
FDMA	Frequency Division Multiple Access
FH	Frequency Hopping
ISMA	Inhibit Sense Multiple Access
IWC	Indoor Wireless Communications
LOS	Line Of Sight
PDF	Probability Density Function

PN	Pseudo Noise
PRMA	Packet Reservation Multiple Access
PSD	Program Structure Diagram
RF	Radio Frequency
SDL	Specification and Description Language
SIR	Signal to Interference Ratio
TDMA	Time Division Multiple Access
TH	Time Hopping
TP	Transmission Period
IP	Idle Period

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# 1. INTRODUCTION

Indoor Wireless Communication (IWC) offers companies a flexibility not available in the fixed wired systems. For example it is not necessary to bring the network down for several minutes to add, remove or move the terminals to other places.

Suppose the company employs a group of representatives, who are most of the time in 'the field' for negotiation with customers and armed with powerful portable computers. When the representatives are back in the office, they can join in the network any time they like it to. In case of the fixed wired systems, it is necessary to perform a routine exercise like plugging and unplugging which can cause network failures because of bad connections. If the computers can communicate with each other wirelessly, they can participate in the network without problems concerning network connections. Time-consuming activities like rearranging the office to account for network connections is history. We can conclude that IWC offer increased flexibility and efficiency, reduction of wiring and down time of services and time and cost saving. Before we can make use of these advantages, we have to solve the problems of the scarce electromagnetic spectrum and the hostile transmission medium first.

Because the electromagnetic spectrum is scarce, it must be used as efficiently as possible. Therefore, it is necessary to solve the problem of the scarce bandwidth. Universities, research laboratories and industries are putting forward enormous effort to investigate new system concepts that can improve the spectrum efficiency. A very important subject with respect to the spectrum efficiency is the choice of a multiple access technique.

Indoor wireless office communication is our main research field. The office system we focus on consists of a building in which users work together in groups. The participants generate terminal traffic. Terminals communicate with each other by radio transmission using a random access protocol. Terminals might not detect each other's transmission in radio communications. It can easily happen that two users are hidden from each other by some obstacle, in which case they are not aware of each other's transmission and collisions can occur. This is called the *hidden terminal problem*. This results in severe performance degradation. The introduction of a central base station can alleviate this problem by instructing it to send a busy tone to all participating terminals to forbid new transmissions

when a transmission is going on. Still, a situation can occur in which two or more terminals simultaneously start their transmission, resulting in a collision. However, a great reduction in the number of simultaneous transmissions is achieved by the introduction of a central base station. This concept is called ISMA [1]-[4].

If we could somehow increase the survival chances of colliding packets, we could improve protocol performance. The near-far effect is one way to achieve this. A packet may 'capture' the receiver if its received power is much stronger than its competitors. This can happen when terminals use the same transmission power, but are located at different distances from the receiver. Because the 'near terminals' have better performance compared to the 'far terminals' (due to the better survival chance of the packets), the near-far effect introduces an unfair element among the terminals. Perfect power control, in which the transmitted power of the terminals are adjusted such that their received powers are all equal, can eliminate the near-far effect described above. To increase the survival chances of colliding packets, we use the CDMA concept. Herefore, each terminal multiplies a specific code sequence to the data sequence before transmission. The receiver uses the same code sequence to gain back the data sequence. The receiver can lock into the first incoming packet and receive this packet correctly even when collisions occur. The performance of CDMA has been reported in a number of publications, e.g. [5]-[10].

The hybrid CDMA/ISMA protocol combines the advantages of both CDMA and ISMA into one protocol. The advantages of CDMA and ISMA are the improvement of the survival chance of data packets and the limitation of contention in the channel. Code sharing can also be applied. This is an important aspect because the number of distinct useful transmission codes is limited, especially for short code length.

In [11] and [12], the hybrid protocol combines Direct Sequence CDMA with *slotted* p-persistent ISMA. In this report, we investigate the performance of the hybrid CDMA/ISMA protocol using the *unslotted* non-persistent ISMA scheme. It is expected that for low traffic, the delay of the unslotted protocol improves because when a terminal has a data packet to send, it does not have to wait until the start of the next time slot. In addition, even when data packets collide, there is a probability that the data is received correctly due to the use of CDMA. This is the strength of the hybrid CDMA/ISMA protocol. By using the *unslotted non-persistent* ISMA scheme, it is expected that we take more advantage of this strength.

Fading is a result of the propagation of the transmitted signal through several paths. In this report, we model the channel as a Rayleigh fading channel. This assumption is valid if the received signal is composed of faded paths with more or less equal power. This means that a line of sight path is absent. For signal modulation, we employ Differential Phase Shift Keying (DPSK).

The performance of the hybrid CDMA/ISMA protocol is measured in terms of throughput and delay in relationship with the offered traffic. To this end, we follow two ways to derive the protocol's performance. First, a computer simulation program is set up. Second, we develop closed form formulas by mathematical analysis. To facilitate the calculations, we adopt a symmetrical star-connected network for our analysis.

This report is organised as follows. Chapter 2 gives the system description. The derivation of the DPSK bit error rate in Rayleigh fading channel is then given in Chapter 3. Chapter 4 describes the performance analysis of the hybrid protocol by simulation. The mathematical analysis then follows in Chapter 5. Subsequently, in Chapter 6, the results will be shown. Finally, conclusions and recommendations are drawn in Chapter 7.



## 2. SYSTEM DESCRIPTION

This chapter introduces the system description as will be used in the analysis. Special attention will be given to CDMA and ISMA, the basic schemes for the hybrid CDMA/ISMA protocol. Also important assumptions with respect to the hybrid CDMA/ISMA protocol will be discussed. This chapter starts with a general discussion of the different types of multiple access protocols.

### 2.1 Multiple Access Protocols

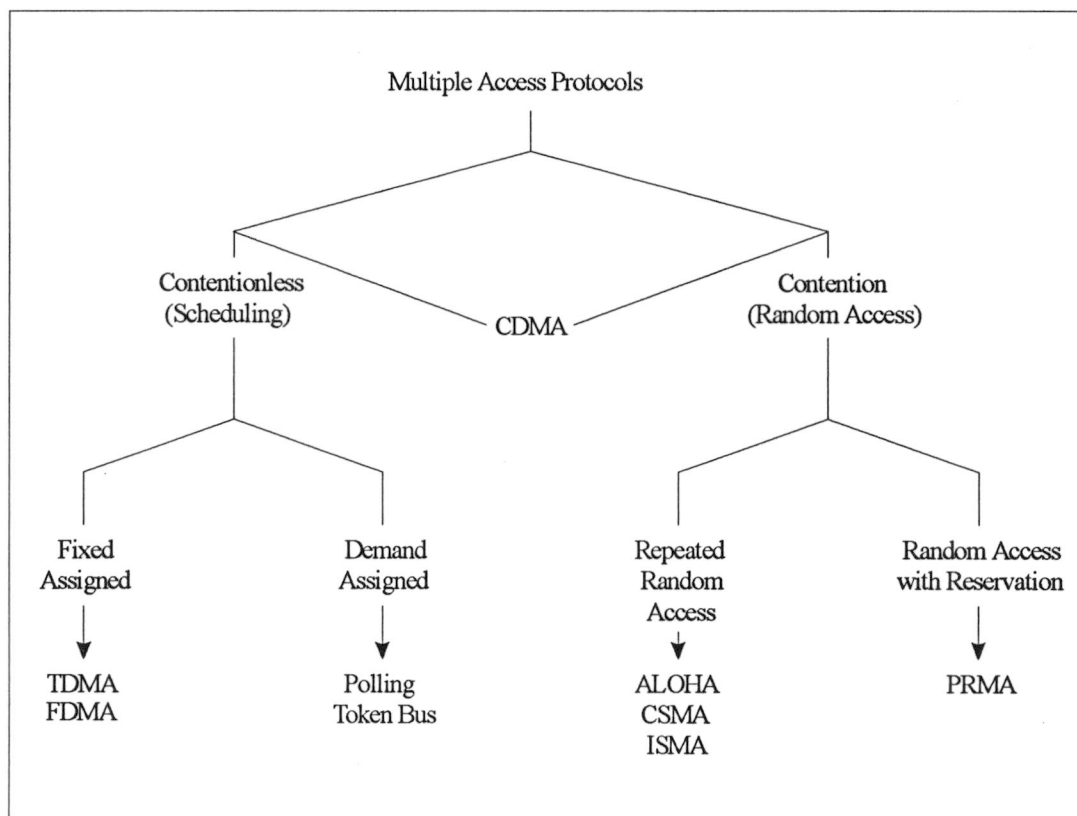


Fig. 2.1: Classification of the multiple access protocols

Fig. 2.1 depicts the classification of the multiple access protocols. Multiple access protocols can be divided in two classes, *contentionless* (or *scheduling*) and *contention* (or *random*

access) [13]. The contentionless class of protocols do not allow contention in the channel. The users are scheduled to transmit in an orderly manner. On the other hand, the contention class does allow contention in the channel. For this class of protocol, collisions can occur because users may access the channel at the same time. CDMA is a protocol which belongs to both classes. This protocol will be described in section 2.3.

### 2.1.1 Contentionless Multiple Access Scheme

Among the class of contentionless we reckon protocols which avoid the simultaneous access of the channel by two or more users. There are two ways to do this. The first *fixed assigned* method assigns to each user a fix amount of the channel's capacity, independent of his activity. If a user is not active, then his capacity is wasted because it can not be used by others. Time Division Multiple Access (TDMA) and Frequency Division Multiple Access (FDMA) belong to this method. *Demand assigned* is the second method in the class of contentionless multiple access protocols. This method only assigns channel's capacity to a user if he has something to transmit. We consider polling and token bus to be demand assigned methods.

### 2.1.2 Contention Multiple Access Scheme

In contrast to the contentionless multiple access scheme where transmissions are organised orderly, transmissions of the class of contention multiple access protocols are not scheduled in an orderly manner. A user does not know about the intention of other users. At most, an ongoing transmission can be detected. This class of protocols can be further subdivided into two groups, *repeated random access* and *random access with reservation*. The repeated random access method uses retransmissions to resolve conflicts. We consider ALOHA, Carrier Sense Multiple Access (CSMA) and Inhibit Sense Multiple Access (ISMA) to belong to this class of protocols. With the random access with reservation scheme, only the first transmission suffers from contention. Once a user has access to the channel, capacity is reserved to this user and he is the only one who may use this. If this user is idle for some

time, then the capacity is reclaimed. Packet Reservation Multiple Access (PRMA) belongs to this type of protocol.

## 2.2 Fixed Assigned Contentionless Multiple Access Protocols

In this section, the fixed assigned contentionless multiple access protocols (TDMA and FDMA) will be discussed in more detail.

### 2.2.1 Frequency Division Multiple Access

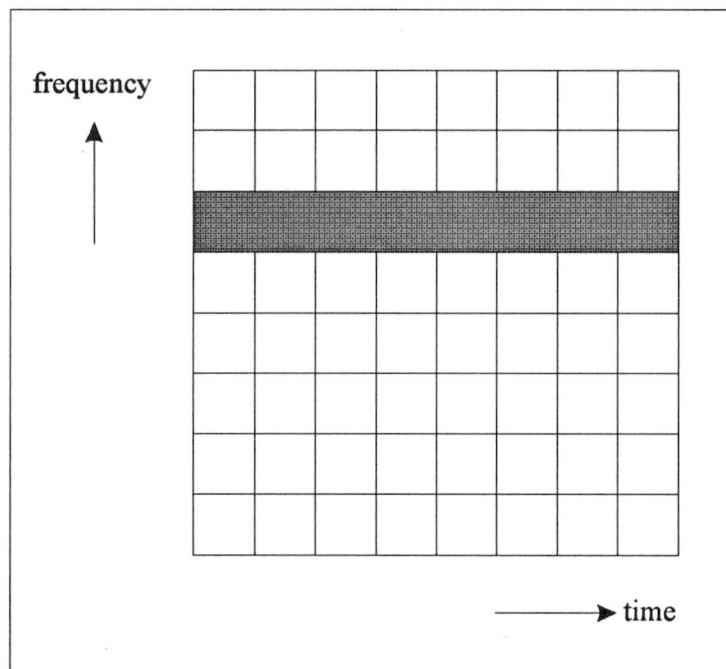


Fig. 2.2: Principle of FDMA

The classical method of providing multiple access capability is FDMA. In FDMA, each user is assigned a particular frequency band. An assigned frequency band can only be used by a particular user all the time and everybody else is excluded from this band (Fig. 2.2).

When all frequency bands are occupied, the system has reached its capacity and no further users may be added.



### 2.2.2 Time Division Multiple Access

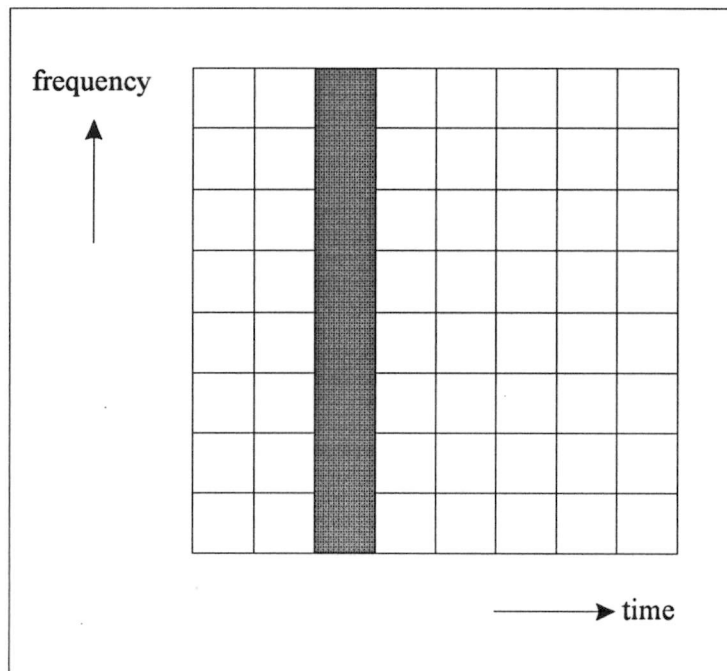


Fig. 2.3: Principle of TDMA

A more recent technique for providing multiple access is TDMA. In TDMA, the time is divided into equal frames and the frames on their turn are divided into equal time slots. Each user is assigned a particular time slot within a time frame, and during

this time slot, transmits a portion of a message by any standard digital technique. Each user is the sole 'owner' of his particular assigned time slot (Fig. 2.3). Time-division multiple access is an important application for satellite communications as well as ground-based digital communication systems. Again, however, when all time slots are occupied, the system is operating at capacity and no more users can be added.

## 2.3 Code Division Multiple Access

Code division multiple access (CDMA) is always accomplished by means of spread spectrum. In this system each user is assigned a particular code. Each user employs his own code to spread the signal into a wide band signal. Fig. 2.4 shows the transformation of the original data signal into a spread spectrum signal. The horizontal axis represents the frequency while the vertical axis depicts the power spectral density. At the receiver side, the same code is used to transform the wide band signal back to the original signal. The cross

correlation between users can be decreased by a proper choice of codes. If we manage to realise a low cross correlation, then we can bring down the interference between users [14]. In contrast to the previously mentioned methods of multiple access, code-division multiple access does not have any sharply defined system capacity. As the number of users increases, the signal-to-interference ratio (SIR) becomes smaller and there is a gradual degradation in performance until the SIR falls below a certain threshold. Thus the system can tolerate significant amounts of overload if the users are willing to tolerate poorer performance.

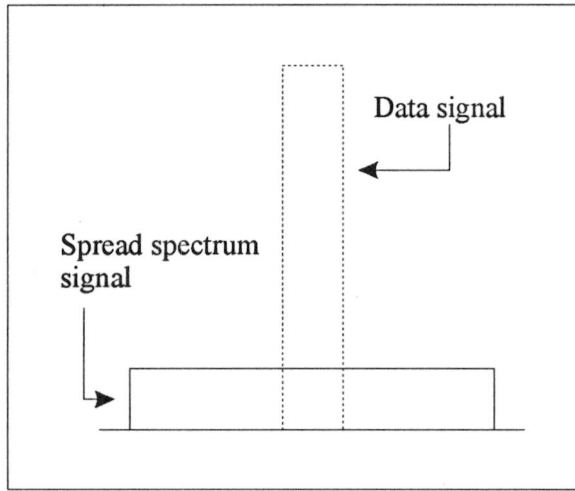


Fig. 2.4: Spreading of the data signal

An additional advantage of CDMA is that the messages intended for one user are not readily decodable by other users, because they may not know the proper codes or have the equipment for generating the appropriate reference signals. Thus there is a privacy feature that is not available in other multiple-access techniques.

The Shannon formula, which express the channel capacity, is the basis of spread spectrum technology:

$$C = W \cdot \log_2 \left( 1 + \frac{S}{N} \right) \quad (2.1)$$

where  $C$  = Capacity in bits per second,

$W$  = Bandwidth in hertz,

$S$  = Signal power,

$N$  = Noise power.

Equation (2.1) shows the trade-off between the transmitted power and the signal bandwidth. Spread spectrum is based on the increase of the signal bandwidth, thereby allowing the signal to noise ratio to decrease to achieve a certain channel capacity.

### 2.3.1 Processing Gain

The most commonly used quantity in describing or specifying spread spectrum systems is that of processing gain ( $G_p$ ):

$$G_p = \frac{BW_{RF}}{R_{data}} \quad (2.2)$$

Where the RF bandwidth ( $BW_{RF}$ ) is the bandwidth of the transmitted spread spectrum signal and the information rate ( $R_{data}$ ) is the data rate in the information baseband channel. So, the ratio between transmitted and original bandwidth is defined as the processing gain.

### 2.3.2 Properties of Spread Spectrum

An application of spread spectrum that is of particular interest in mobile communications is the ability of a wide-band signal to resist the effects of multipath fading. A property of multipath fading is that frequencies separated by only a few hundred kilohertz may fade essentially independently. Thus at any given time when the signal has a large bandwidth, only a small portion of the bandwidth will be in a fade. The average received signal power thus can be made more nearly constant than it would be for a narrow-band signal. This resistance to fading is an important consideration in the potential application of spread spectrum to mobile communication situations.

Another important element employed in the design of spread spectrum signals is pseudo-randomness. It makes the signals appear similar to random noise and difficult to demodulate by receivers other than the intended ones. A message may be hidden in the background noise by spreading its bandwidth with coding and transmitting the resultant

signal at a low average power. Because of its low power level, the transmitted signal is said to be 'covert'. It has a low probability of being intercepted (detected) by a casual listener.

Message privacy may be obtained by superimposing a pseudo-random pattern on a transmitted message. The message can be demodulated by the intended receivers that know the pseudo-random pattern or key used at the transmitter but not by any other receivers that do not have knowledge of the key. Resuming the advantages spread spectrum offers we come to the following:

- Interference rejection;
- Anti-multipath fading (frequency diversity);
- Low probability of intercept;
- Secrecy;
- Multiple access capability.

### 2.3.3 CDMA Techniques

We can distinguish between three main types of spread spectrum techniques: Direct Sequence (DS), Frequency Hopping (FH) and Time Hopping (TH).

For the direct sequence CDMA method, a digital information stream is modulated by a binary code sequence (Fig. 2.5). The recovering of the signal at the receiver is done by correlating the received signal with the original code sequence used by the transmitter. In a frequency hopped spread spectrum communications system, the available channel bandwidth is subdivided into a large number of adjacent frequency bands. In any signalling interval, the transmitted signal occupies one or more of the available frequency bands. The selection of the frequency band(s) in each signalling interval is made pseudo-randomly according to the output from a pseudo noise (PN) generator. For the time hopping method, the time axis is divided into frames, which in turn are subdivided into time slots. The user transmits the data in a different time slot every frame, using all the available bandwidth.

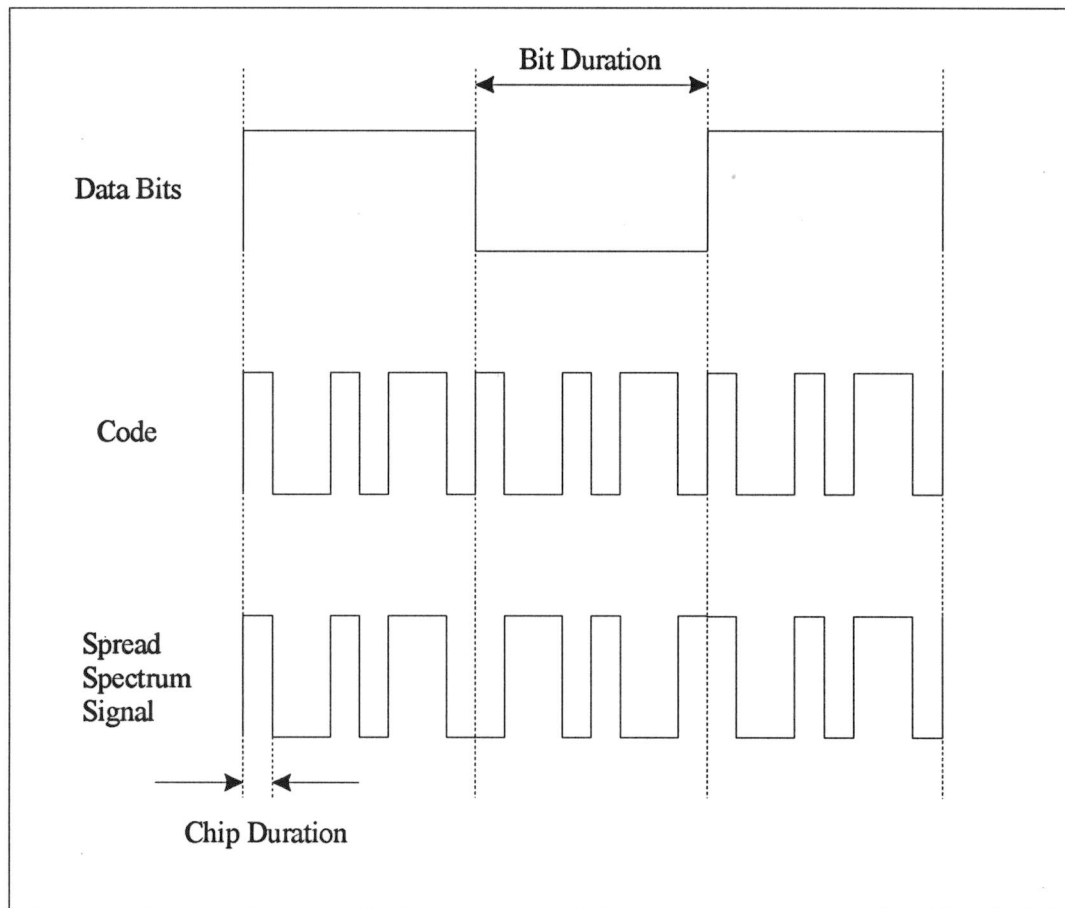


Fig. 2.5: Direct Sequence CDMA signal spreading

## 2.4 Repeated Random Access Protocols

ALOHA, CSMA and ISMA schemes belong to this type of repeated random access protocols. We only discuss the slotted p-persistent ISMA and the unslotted non-persistent ISMA protocol in this section.

### 2.4.1 The Slotted P-Persistent ISMA Protocol

The Specification and Description Language (SDL) diagram of the slotted p-persistent ISMA protocol is depicted in Fig. 2.6. Before transmitting the data packet to the receiver,

the terminal first listens to the channel to detect whether there is a busy tone transmitted by the base station. The base station broadcasts a busy tone to indicate that the channel is busy because there is a transmission going on.

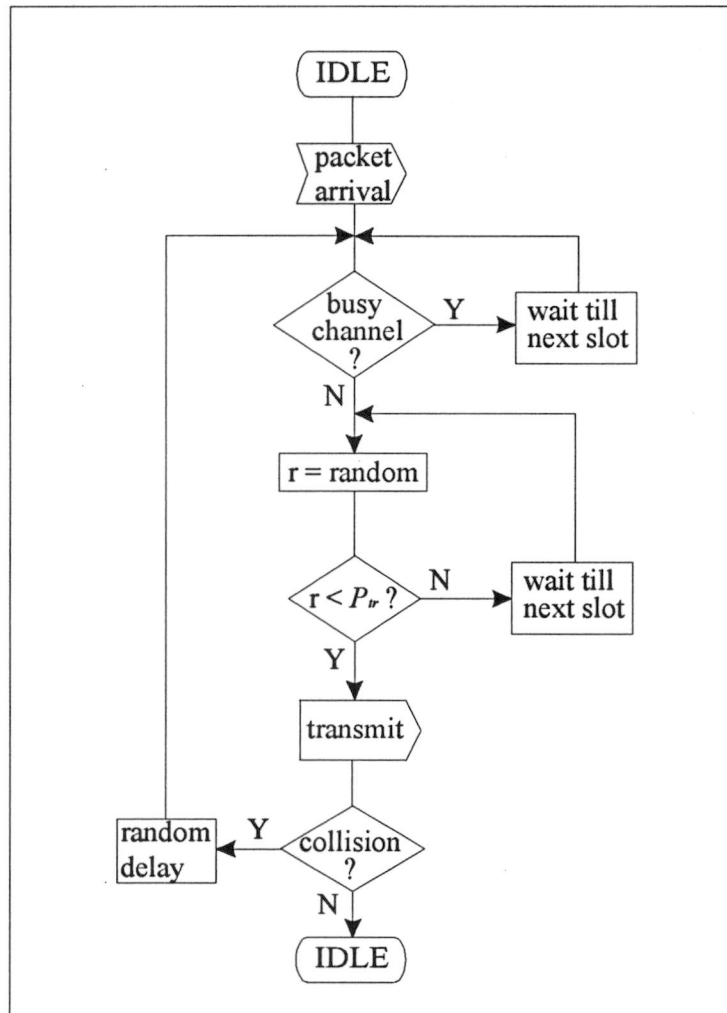


Fig. 2.6: Slotted p-persistent ISMA protocol

time slot and draws a number again to see if it's permitted to send. This mechanism is built in to reduce the number of simultaneous transmissions.

If the terminal is permitted to send, it transmits its data packet to the receiver. It is possible that two or more terminals transmit in the same time slot. In this case a collision occurs and the data packets are damaged. The terminals then wait a random delay and start all over again (see also Fig. 2.6).

If there is a busy tone, indicating that the channel is busy, the terminal postpones its transmission to the next time slot. Otherwise, if the channel is free, the terminal draws a number to see if it is permitted to send its data packet. The probability that a terminal succeeds is  $P_{tr}$ . (If  $P_{tr}=0.1$  then the protocol is called 0.1-persistent ISMA). In case the terminal is unlucky, it waits till the next

### 2.4.2 The Unslotted Non-Persistent ISMA Protocol

The SDL-diagram of the unslotted non-persistent ISMA protocol is shown in Fig. 2.7. This ISMA protocol is unslotted. There are two major differences between this unslotted and the previously in section 2.4.1 described slotted protocol.

The first difference is that the terminal waits a random delay when it senses that the channel is busy. Because the time axis is not divided into time slots, a random delay has to be introduced. After the random delay, terminals are allowed to sense the channel again to see if it's free.

The second difference is related to the first one. In the unslotted scheme, the terminals immediately transmit if the channel is free. The terminals already wait a random delay in case the channel is busy. So, if the channel is free, the number of terminals attempting to transmit is already reduced. It is not necessary to let the terminals draw a random number again to see if they may transmit.

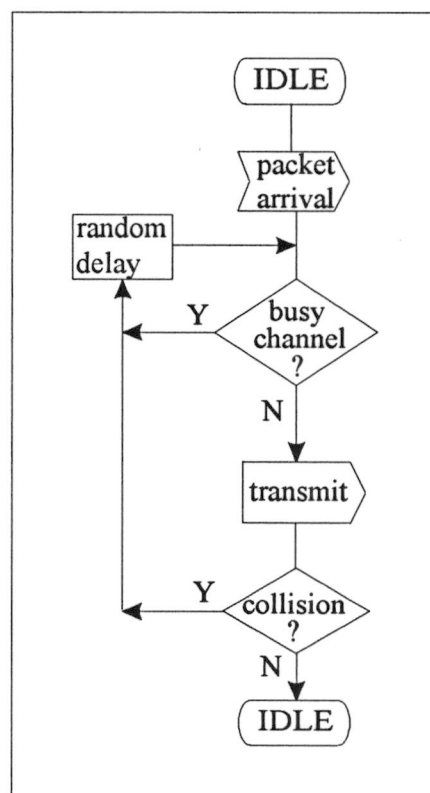


Fig. 2.7: Unslotted non-persistent ISMA

Although the number of collisions is greatly reduced with ISMA, collisions still can occur. In this unslotted ISMA scheme, collisions may occur due to multiple terminals transmitting data packets during an interval called the *inhibit delay fraction*  $d$ . This interval  $d$  is necessary to switch the busy tone from 'idle' to 'busy'. The reverse interval from 'busy' to 'idle' is denoted by  $d'$ . The inhibit delay fraction is normalised to the packet length, resulting in  $0 \leq d < 1$  (Fig. 2.8). In this report we assume that  $d' = d$ .

The analysis of the hybrid protocol using slotted p-persistent ISMA has already been performed ([11] and [12]). We now concentrate on the hybrid protocol using the unslotted

non-persistent ISMA scheme. In case of the unslotted protocol, we expect that for low traffic, the average packet delay can be reduced because terminals do not have to wait to the start of the new time slot.

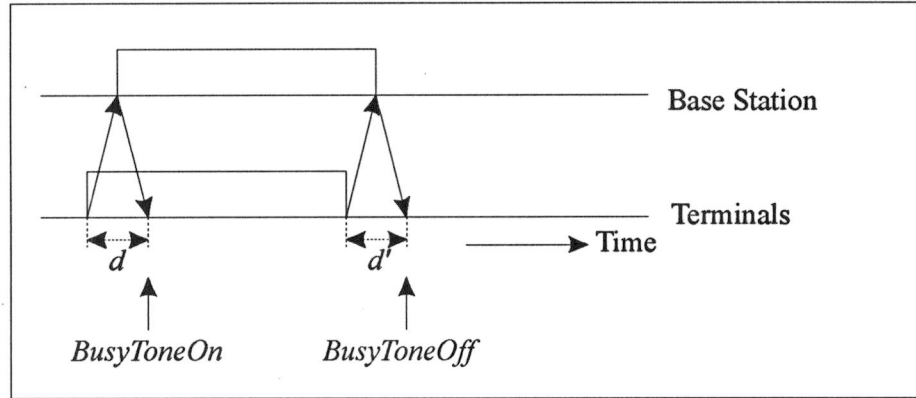


Fig. 2.8: Inhibit delay fraction  $d$

## 2.5 Description of the Hybrid CDMA/ISMA Protocol

Users behind terminals generate data traffic, which is divided in data packets. The data packets arrive at terminals, who take care of the correct delivery. We assume that the arrivals of these packets are generated by a Poisson process. The state of the terminals can either be free or blocked. At the start, the terminals are in the free state. If a packet arrives at a free terminal, the terminal jumps into the blocked state. In the blocked state, the terminal takes care that the arriving data packet is serviced successfully. In the mean time, the blocked terminal ignores all incoming packets. This is a consequence of the assumption that the terminals don't have buffers for the incoming data packets. We assume that the layer above will handle this correctly. In the blocked state, the terminal uses the unslotted non-persistent ISMA protocol to service its data packets. If a data packet is received erroneously, then the ISMA scheme takes care that this problem is solved. Only when the data packet is received correctly, the terminal leaves the blocked state to jump into the free state. New data packets can then be handled.



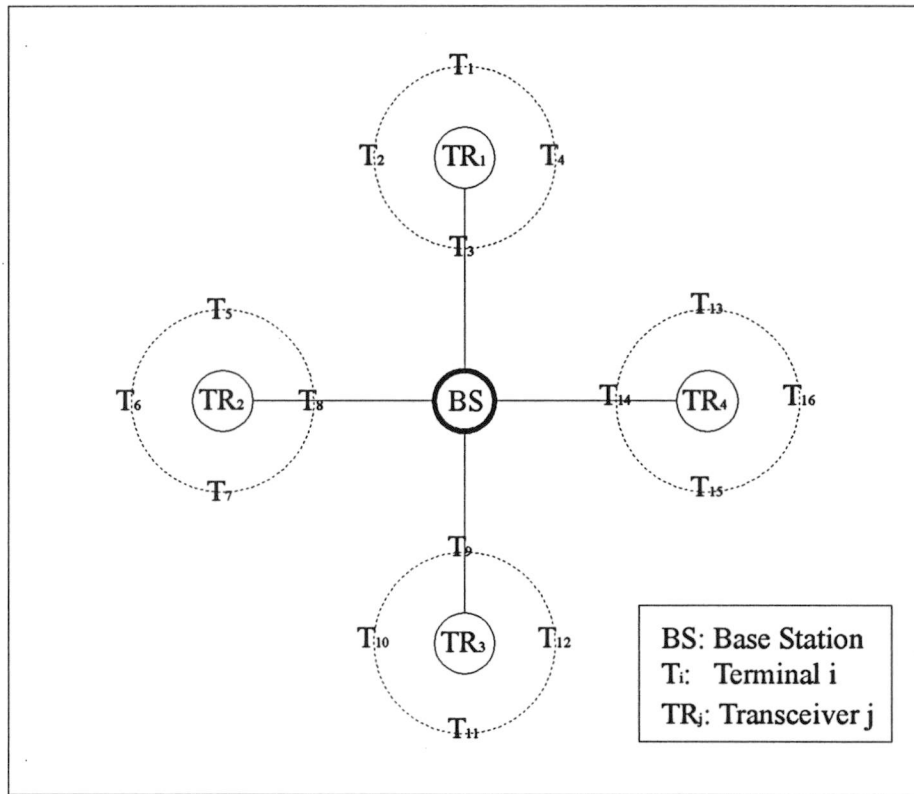


Fig. 2.9: 16-terminal network configuration

Fig. 2.9 depicts an example of the 16-terminal network configuration of the protocol. There is one central base station that is connected to several transceivers by wire. The base station controls the traffic flow with a busy tone that can be detected by all participated terminals. Because all terminals can detect the busy tone, the hidden terminal problem is solved. Several terminals together with one transceiver form a group. Wireless communication is considered to take place between the transceiver and the terminals. The terminals around each transceiver share the same code. In this way the number of codes can be reduced. When a data packet is ready for transmission, a terminal transmits the packet to its transceiver according to the chosen ISMA scheme.

After the arrival of the data packet, the transceiver then simply forwards this packet to the base station which forwards it to all terminals via the transceivers. The destination then receives the packet and decides whether the packet is errorless or not. In case the packet was erroneous, a retransmission must take place. The return channel is not included in our analysis because only the base station makes use of this return channel and therefore contention is not a concern.

## 2.6 Assumptions

It is impossible and mostly also not necessary to take all complexities of the real life system into account. That is why assumptions have to be made in order to be able to analyse a certain aspect of the system. It is also important to list the assumptions used to indicate under which conditions the generated result is valid. The assumptions we adopted in our work are listed here below. Unless stated otherwise, the values of the parameters we use to obtain the protocol's performance is also given in this section.

### 2.6.1 Arrival Process

The arrival of data packets at terminals is assumed to be generated by a Poisson process. This means that the inter-arrival times of the data packets at terminals are negative exponentially distributed with  $\lambda$  as parameter. Before a retransmission (caused by a collision or a busy channel) can take place, terminals have to undergo a random delay first (see Fig. 2.7). This random delay is also negative exponentially distributed with  $\alpha$  as parameter.  $\alpha$  is chosen to be 0.1 and the inhibit delay fraction  $d$  is 0.01. The negative exponential distribution will be discussed in Chapter 4.

### 2.6.2 Signal and Channel Characteristics

After Differential Phase Shift Keying (DPSK) modulation, the signal is spread with a Gold code. The length of this code is chosen to be 31 or 127. The signal is then transmitted, modulated on a 1.7 GHz carrier. The data rate is arbitrarily chosen to be 256·1024 b/s (0.26 Mb/s). The packet length is 64 bits. The delay spread is supposed to be 100 ns and the SNR is 20 dB. Perfect power control is assumed to assure that signals from terminals within the same group arrive at the transceiver with the same power.

The far field model [15] is chosen to describe the signal attenuation. Given the transmitted power  $P_T$ , the received power  $P_R$  can be expressed by the following equation:

$$P_R = \frac{g_T g_R}{\left(\frac{4\pi fl}{c}\right)^\mu} P_T \quad (2.3)$$

where  $g_T$  and  $g_R$  are respectively the transmitter and the receiver gains,  $f$  the signal frequency,  $l$  the distance between the transmitter and the receiver,  $c$  the speed of light and  $\mu$  is the attenuation parameter. This attenuation parameter is chosen equal to 2, corresponding to free space propagation.

Fading is the result of the propagation of the transmitted signal through several paths. The channel is modelled as a Rayleigh fading channel. The Rayleigh fading channel model is valid when each path contributes the same amount of energy to the composite received signal. A Line of Sight (LOS) path is therefore assumed to be absent.

### 2.6.3 Network Configuration

An example of a 16-terminal network configuration that we adopted in this paper is shown in Fig. 2.9. The terminals are clustered around distributed transceivers at a fixed distance (5 m). The distributed transceivers are clustered around the base station at a fixed distance (30 m). Those distances are kept fixed throughout the analysis. The number of terminals within a transceiver group (also called group size) is chosen as a power of two. For a certain fixed number of participated terminals, we have to halve the group size if we want to double the number of codes (the number of codes also equals the number of transceivers). The comparison is fair in this manner, because when we want to investigate the effect of the number of codes on the performance, we have to keep the number of participating terminals fixed. The number of terminals is 32 unless stated otherwise.

### 3. BIT ERROR RATE IN RAYLEIGH FADING CHANNEL

The bit error rate for direct-sequence spread spectrum with Differential Phase Shift Keying (DPSK) modulation and diversity in Rayleigh fading channel is evaluated in this chapter. The transmitter, channel and receiver model will be presented in the subsequent sections and are similar to those in [16].

#### 3.1 Transmitter Model

The number of active users in the indoor wireless communications system is denoted by  $K$ . Fig. 3.1 depicts the CDMA transmission model. User  $i$  applies code  $i$  to communicate with the base station.

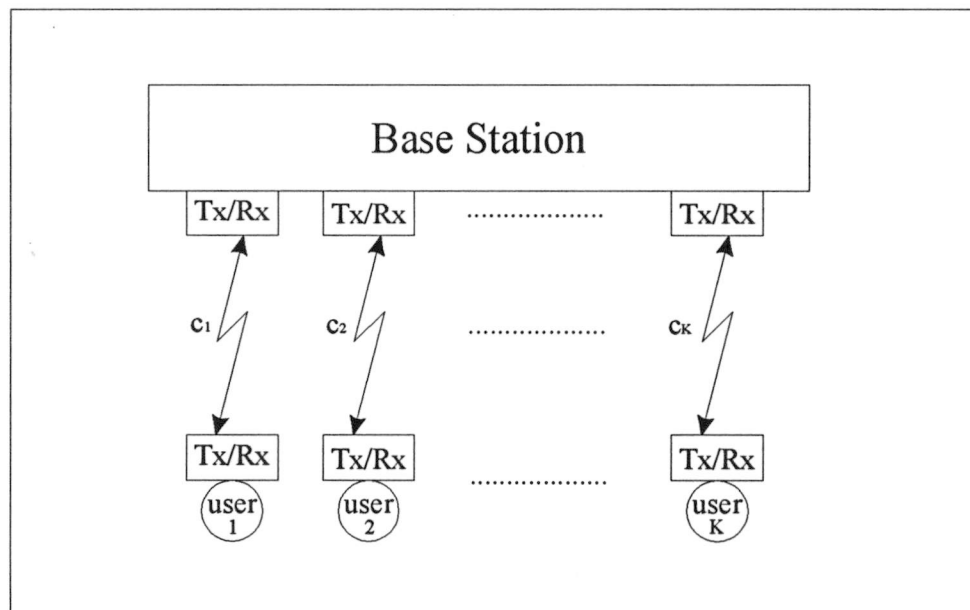


Fig. 3.1: CDMA transmission channel

The  $k$ th user's information signal  $b_k$  is the differentially encoded binary data sequence of rectangular pulses of width  $T_b$ . To spread each data bit, a code waveform  $a_k$  of  $N$  rectangular pulses of width  $T_c$  is used. In literature,  $a_k$  is also called the chip waveform. There is one period of code sequence per data bit resulting in  $T_b = NT_c$ . The chip and data waveforms for the  $k$ th user are denoted by  $a_k(t)$  and  $b_k(t)$  respectively and are given by:

$$a_k(t) = \sum_i a_k^i P_{T_c}(t - iT_c), \quad a_k^i \in \{-1, 1\} \quad (3.1)$$

$$b_k(t) = \sum_{j=-\infty}^{\infty} b_k^j P_{T_b}(t - jT_b), \quad b_k^j \in \{-1, 1\} \quad (3.2)$$

Here  $a_k^i$  represents the  $i$ th chip value of the  $k$ th user,  $b_k^j$  the  $k$ th user data bit at the  $j$ th timing interval and  $a_k^i = a_k^{i+N}$  for all  $i$ ;  $P_W(t)$  is a rectangular pulse of unit height and width  $W$ .

Spreading is achieved by multiplying (or modulo-2 adding) the direct-sequence code to the data signal. The spread signal is then modulated onto the RF carrier signal  $A \cos(\omega_c t + \theta_k)$ . The carrier frequency is the same for all users and is denoted by  $\omega_c$ , while the carrier phase for the  $k$ th user is  $\theta_k$  and  $A$  is the carrier level (the energy per bit  $E_b$  equals  $A^2 T_b / 2$ ). It is further assumed that  $\omega_c T_b = 2l\pi$ , in which  $l$  is an integer. The transmitted signal for the  $k$ th user  $s_k(t)$  is given by:

$$s_k(t) = A a_k(t) b_k(t) \cos(\omega_c t + \theta_k) \quad (3.3)$$

## 3.2 Channel Model

The emitted information reaches the destination not only via the line of sight (LOS) path but also via reflections by buildings and walls. In case the line of sight path dominates the other reflected signals, the sum of the signals at the receiver can be modeled by the Rician

distribution. When all the signals are received with approximately the same power, the receiver power distribution can be modeled by the Rayleigh distribution (Fig. 3.2).

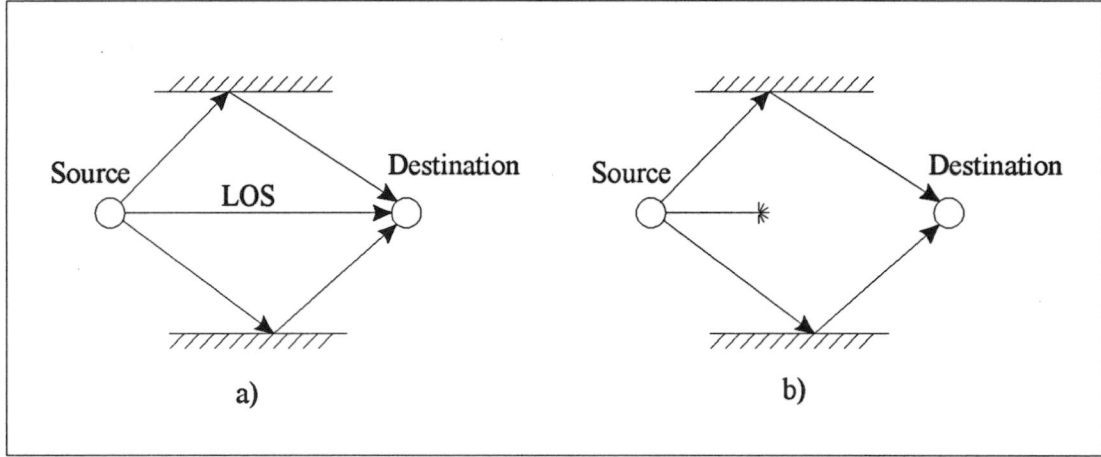


Fig 3.2: Fading channels a) Rice; b) Rayleigh

For our work we model our channel as a Rayleigh fading channel. The complex lowpass equivalent impulse response of the multipath Rayleigh fading channel for the link between the  $k$ th user and the base station (in our case it is the transceiver) is given by:

$$h_k(\tau) = \sum_{l=1}^L \beta_{lk} \delta(\tau - \tau_{lk}) e^{j\gamma_{lk}} \quad (3.4)$$

where  $\beta$  is the path gain and is Rayleigh distributed,  $\tau$  is the path delay and  $\gamma$  is the path phase. The index  $lk$  refers to the  $l$ th path of the  $k$ th user, and  $j^2 = -1$ .  $\delta(\cdot)$  is the Dirac delta function and  $L$  is the number of resolvable paths and is upper bounded by:

$$L = \left\lfloor \frac{T_m}{T_c} \right\rfloor + 1 \quad (3.5)$$

In which  $\lfloor x \rfloor$  is the largest integer smaller than or equal to  $x$  [12],  $T_m$  is the rms delay spread and depends on the size and type of buildings. Reported values are between 20 and 50 ns for small and medium-size office buildings, between 30 and 300 ns for various factory environments, under 100 ns at several university buildings, less than 160 ns over 90% of the

area in a shielded building, less than 80 ns in an office building, under 120 ns in a large office building, and up to 200 ns in other large office buildings [17]. For our work we adopted 100 ns as the value for  $T_m$ .

We assume that the path delays  $\tau_{lk}$  are independent random variables and uniformly distributed over  $[0, T_b]$ . Fading is the result of the propagation of the transmitted signal through several paths. The channel is modeled as a Rayleigh fading channel. The Rayleigh fading channel model is valid when each path contributes the same amount of energy to the composite received signal. A Line of Sight (LOS) path is therefore assumed to be absent (Fig. 3.2). So, the path gain  $\beta_{lk}$  is an independent Rayleigh random variable for each  $l$  and  $k$ . The Rayleigh distribution for  $\beta_{lk}$  is given as:

$$P_{\beta_{lk}}(r) = \frac{r}{\rho_{lk}} \exp\left(-\frac{r^2}{2\rho_{lk}}\right), \quad r \geq 0 \quad (3.6)$$

In which  $r$  is the signal amplitude and  $\rho_{lk} = \frac{1}{2}E[\beta_{lk}^2]$  is the average scattered power.

### 3.3 Receiver Model

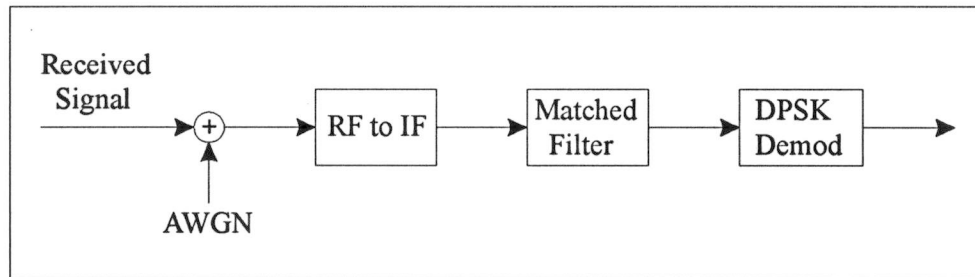


Fig. 3.3: Spread spectrum receiver using DPSK modulation

The receiver consists of a RF to IF converter, a matched filter and a DPSK demodulator (Fig. 3.3). We assume that the received signal  $r(t)$  is composed of the contributions of the different users, their different paths and additive white Gaussian noise (AWGN). The receiver input signal at the antenna for a certain user can be written as:

$$r(t) = \sum_{k=1}^K \sum_{l=1}^L A\beta_{lk} a_k(t - \tau_{lk}) b_k(t - \tau_{lk}) \cos(\omega_c t + \phi_{lk}) + n(t) \quad (3.7)$$

Here,  $n(t)$  is the white Gaussian noise with two-sided power spectral density  $N_0/2$  [W/Hz]. We assume that the path phase  $\phi_{lk}$ , given by  $(\omega_c \tau_{lk} + \gamma_{lk} + \theta_k)$ , is an independent random variable uniformly distributed over  $[0, 2\pi]$ . In terms of low-pass signals,  $r(t)$  and  $n(t)$  can also be expressed as:

$$r(t) = x(t) \cos(\omega_c t) - y(t) \sin(\omega_c t) \quad (3.8)$$

$$n(t) = n_c(t) \cos(\omega_c t) - n_s(t) \sin(\omega_c t) \quad (3.9)$$

Here,  $n_c(t)$  and  $n_s(t)$  are the lowpass equivalent components of  $n(t)$  and

$$x(t) = \sum_{k=1}^K \sum_{l=1}^L A\beta_{lk} a_k(t - \tau_{lk}) b_k(t - \tau_{lk}) \cos(\phi_{lk}) + n_c(t) \quad (3.10)$$

$$y(t) = \sum_{k=1}^K \sum_{l=1}^L A\beta_{lk} a_k(t - \tau_{lk}) b_k(t - \tau_{lk}) \sin(\phi_{lk}) + n_s(t) \quad (3.11)$$

Selecting user 1 as the reference user, each component of the received signal is multiplied by the direct sequence code associated with user 1. The output (in-phase and quadrature component) of the matched filter of user 1 at the sampling instant  $(t = T_b)$  is then given as:

$$g_x(T_b) = \sum_{k=1}^K \sum_{l=1}^L A\beta_{lk} \cos(\phi_{lk}) \left[ b_k^{-1} R_{1k}(\tau_{lk}) + b_k^0 \hat{R}_{1k}(\tau_{lk}) \right] + \eta \quad (3.12)$$

$$g_y(T_b) = \sum_{k=1}^K \sum_{l=1}^L A\beta_{lk} \sin(\phi_{lk}) \left[ b_k^{-1} R_{1k}(\tau_{lk}) + b_k^0 \hat{R}_{1k}(\tau_{lk}) \right] + \nu \quad (3.13)$$



Where  $g_x$  and  $g_y$  are the in-phase and the quadrature components,  $b_k^{-1}$  the previous and  $b_k^0$  the current data bit. The noise samples  $\eta$  and  $\nu$  are independent, zero-mean Gaussian random variables with identical variance  $\sigma_n^2 = N_0 T_b$ . Further,

$$R_{1k}(\tau) = \int_0^\tau a_k(t - \tau) a_1(t) dt \quad (3.14)$$

$$\hat{R}_{1k}(\tau) = \int_\tau^{T_b} a_k(t - \tau) a_1(t) dt \quad (3.15)$$

Let us assume without loss of generality that the receiver synchronizes to the  $j$ th path of user 1, so that  $\tau_{j1} = 0$  and  $\phi_{j1} = 0$ . All other delay paths constitute interference and are then defined as relative according to this reference. The complex envelope of the signal  $z_1 = g_x(T_b) + jg_y(T_b)$  at the current sampling instant is:

$$\begin{aligned} z_1 = & A\beta_{j1}T_b b_1^0 + A \sum_{k=1}^K (b_k^{-1} X_k + b_k^0 \hat{X}_k) \\ & + jA \sum_{k=1}^K (b_k^{-1} Y_k + b_k^0 \hat{Y}_k) + (\eta_1 + j\nu_1) \end{aligned} \quad (3.16)$$

For the previous sampling instant, we have something similar

$$\begin{aligned} z_{-1} = & A\beta_{j1}T_b b_1^{-1} + A \sum_{k=1}^K (b_k^{-2} X_k + b_k^{-1} \hat{X}_k) \\ & + jA \sum_{k=1}^K (b_k^{-2} Y_k + b_k^{-1} \hat{Y}_k) + (\eta_2 + j\nu_2) \end{aligned} \quad (3.17)$$

in which  $b_k^{-2}$  is the  $k$ th user data bit transmitted 2 bits intervals prior to  $b_k^0$  and  $\eta_1, \eta_2, \nu_1$  and  $\nu_2$  are the noise variables independent of one another. Further,

$$X_1 = \sum_{\substack{l=1 \\ l \neq j}}^L R_{1l}(\tau_{1l}) \beta_{1l} \cos(\phi_{1l}) \quad (3.18)$$

$$\hat{X}_1 = \sum_{\substack{l=1 \\ l \neq j}}^L \hat{R}_{1l}(\tau_{1l}) \beta_{1l} \cos(\phi_{1l}) \quad (3.19)$$

$$Y_1 = \sum_{\substack{l=1 \\ l \neq j}}^L R_{1l}(\tau_{1l}) \beta_{1l} \sin(\phi_{1l}) \quad (3.20)$$

$$\hat{Y}_1 = \sum_{\substack{l=1 \\ l \neq j}}^L \hat{R}_{1l}(\tau_{1l}) \beta_{1l} \sin(\phi_{1l}) \quad (3.21)$$

For  $k \geq 2$ ,

$$X_k = \sum_{l=1}^L R_{lk}(\tau_{lk}) \beta_{lk} \cos(\phi_{lk}) \quad (3.22)$$

$$\hat{X}_k = \sum_{l=1}^L \hat{R}_{lk}(\tau_{lk}) \beta_{lk} \cos(\phi_{lk}) \quad (3.23)$$

$$Y_k = \sum_{l=1}^L R_{lk}(\tau_{lk}) \beta_{lk} \sin(\phi_{lk}) \quad (3.24)$$

$$\hat{Y}_k = \sum_{l=1}^L \hat{R}_{lk}(\tau_{lk}) \beta_{lk} \sin(\phi_{lk}) \quad (3.25)$$

The decision variable for DPSK demodulation is

$$\xi = \text{Re}[z_1 z_{-1}^*] \quad (3.26)$$

In which  $\text{Re}[x]$  denotes the real part of  $x$  and  $*$  the complex conjugation. Errors occur if the channel attenuation is large. If we somehow provide the receiver with several independently fading signal paths, the probability that all the signal components will fade simultaneously is

reduced considerably. This is the basis for diversity techniques. The selection diversity of order  $M$  is based on selecting the strongest of the decision variable:

$$\xi_{max}^M = \max_{i=1,\dots,M}(\xi_i) \quad (3.27)$$

By using multiple antennas, the highest possible order of diversity, i.e. number of paths to choose from, can be increased to  $M_{max}=K_aL$  where  $M_{max}$  is the maximum order of diversity and  $K_a$  is the number of antennas. Designate  $\xi_{max}$  as the decision variable obtained from demodulation of the strongest path. The bit error probability in case of selection diversity is defined as

$$P_{be} \Delta P_r(\xi_{max} < 0 | b_1^0 b_1^{-1} = 1) \Delta P_r(\xi_{max} > 0 | b_1^0 b_1^{-1} = -1) \quad (3.28)$$

If we assume that all path delays are given and  $\beta_{max}$  is correctly selected, the formula for the bit error probability is given by [16]:

$$P_{be|\beta_{max}\{\tau_{lk}\},L} = Q(a,b) - \frac{1}{2} \left( 1 + \frac{\mu_{12}}{\sqrt{\mu_1\mu_2}} \right) I_0(a,b) \exp\left(-\frac{a^2+b^2}{2}\right) \quad (3.29)$$

where

$Q(a,b)$  is the Marcum Q-function

$$a = \frac{m}{\sqrt{2}} \left| \frac{1}{\sqrt{\mu_1}} - \frac{1}{\sqrt{\mu_2}} \right| \quad (3.30)$$

$$b = \frac{m}{\sqrt{2}} \left| \frac{1}{\sqrt{\mu_1}} + \frac{1}{\sqrt{\mu_2}} \right| \quad (3.31)$$

$$m = A\beta_{max}T_b b_1^0 = A\beta_{max}T_b b_1^{-1} \quad (3.32)$$

$$\begin{aligned}\mu_1 = A^2 E \left[ \sum_{k=1}^K X_k^2 + \hat{X}_k^2 + Y_k^2 + \hat{Y}_k^2 \middle| \{\tau_{lk}\}, L \right] \\ + 2A^2 E \left[ X_1 \hat{X}_1 + Y_1 \hat{Y}_1 \middle| \{\tau_{lk}\}, L \right] + 2\sigma_n^2\end{aligned}\quad (3.33)$$

$$\mu_2 = A^2 E \left[ \sum_{k=1}^K X_k^2 + \hat{X}_k^2 + Y_k^2 + \hat{Y}_k^2 \middle| \{\tau_{lk}\}, L \right] + 2\sigma_n^2 \quad (3.34)$$

$$\mu_{12} = A^2 E \left[ \sum_{k=1}^K (X_k \hat{X}_k + Y_k \hat{Y}_k) + \hat{X}_1^2 + \hat{Y}_1^2 \middle| \{\tau_{lk}\}, L \right] \quad (3.35)$$

The conditional expectations in the above expressions can be evaluated as follows:

$$E \left[ X_1^2 \middle| \{\tau_{lk}\}, L \right] = E \left[ Y_1^2 \middle| \{\tau_{lk}\}, L \right] = \sum_{\substack{l=1 \\ l \neq j}}^L R_{11}^2(\tau_{l1}) \rho_{l1} \quad (3.36)$$

$$E \left[ \hat{X}_1^2 \middle| \{\tau_{lk}\}, L \right] = E \left[ \hat{Y}_1^2 \middle| \{\tau_{lk}\}, L \right] = \sum_{\substack{l=1 \\ l \neq j}}^L \hat{R}_{11}^2(\tau_{l1}) \rho_{l1} \quad (3.37)$$

$$E \left[ X_1 \hat{X}_1 \middle| \{\tau_{lk}\}, L \right] = E \left[ Y_1 \hat{Y}_1 \middle| \{\tau_{lk}\}, L \right] = \sum_{\substack{l=1 \\ l \neq j}}^L R_{11}(\tau_{l1}) \hat{R}_{11}(\tau_{l1}) \rho_{l1} \quad (3.38)$$

$$E \left[ X_k^2 \middle| \{\tau_{lk}\}, L \right] = E \left[ Y_k^2 \middle| \{\tau_{lk}\}, L \right] = \sum_{l=1}^L R_{1k}^2(\tau_{lk}) \rho_{lk} \quad (3.39)$$

$$E \left[ \hat{X}_k^2 \middle| \{\tau_{lk}\}, L \right] = E \left[ \hat{Y}_k^2 \middle| \{\tau_{lk}\}, L \right] = \sum_{l=1}^L \hat{R}_{1k}^2(\tau_{lk}) \rho_{lk} \quad ; k \geq 2 \quad (3.40)$$

$$E \left[ X_k \hat{X}_k \middle| \{\tau_{lk}\}, L \right] = E \left[ Y_k \hat{Y}_k \middle| \{\tau_{lk}\}, L \right] = \sum_{l=1}^L R_{1k}(\tau_{lk}) \hat{R}_{1k}(\tau_{lk}) \rho_{lk} \quad ; k \geq 2 \quad (3.41)$$



## 4. COMPUTER SIMULATION

A simulation program has been created to analyse the protocol behaviour. In this chapter we describe how the computer simulation program was set up to measure the performance of the unslotted hybrid CDMA/ISMA protocol. The primary goal of this simulation program is to see whether the performance of the hybrid protocol using the *slotted p-persistent* scheme on the one hand and the *unslotted non-persistent* scheme on the other, differ from each other, and if so in what way. Knowledge in this matter give us meaningful insight that we can use to design our protocol appropriately. This chapter also includes the generation of Poisson data traffic.

### 4.1 Simulation Techniques

In general, if we want to use a computer simulation to analyse a system, the following steps can be distinguished. The dynamic behaviour of the system is studied by tracing various system states as a function of time and then collecting and analysing the system statistics. The events that change the system state are generated at different points in time, and the passage of time is represented by an internal clock which is incremented and maintained by the simulation program.

The simulation time can be advanced in two ways. The first method is the *interval-oriented simulation* (or the uniform time increment method) where the clock is advanced from time  $t$  to  $(t + \Delta t)$  where  $\Delta t$  is a uniform fixed time increment. Fig. 4.1a depicts this mechanism. The second method is the *event-oriented simulation* (or the variable time increment method) where the clock is incremented from time  $t$  to the next event time  $t'$ , whatever may be the value of  $t'$ . The state changes are made at event time  $t$ , the next event time  $t'$ , and this process is continuously repeated. Thus, only events are represented explicitly in a simulation model and the periods between events are treated as inactive or insignificant and therefore consume no time even though the inter-event activities do consume time in the real world (Fig. 4.1b).

Obviously, method 1 detects the events that occur during the interval  $(t, t + \Delta t)$  only at time  $(t + \Delta t)$ , thereby introducing errors in simulation. Another drawback of this method is that if the interval between two events is very large compared to  $\Delta t$ , then the simulator goes through several unproductive clock increments (during periods of inactivity) and the associated computing effort which will not bring about any change in system states. This fixed time increment method is suitable for the simulation of continuous systems and in particular systems with large numbers of state variables.

The second method involves sorting of event activation times and maintaining an *event list*. In our work we will employ the event-oriented simulation to analyse the unslotted hybrid CDMA/ISMA protocol.

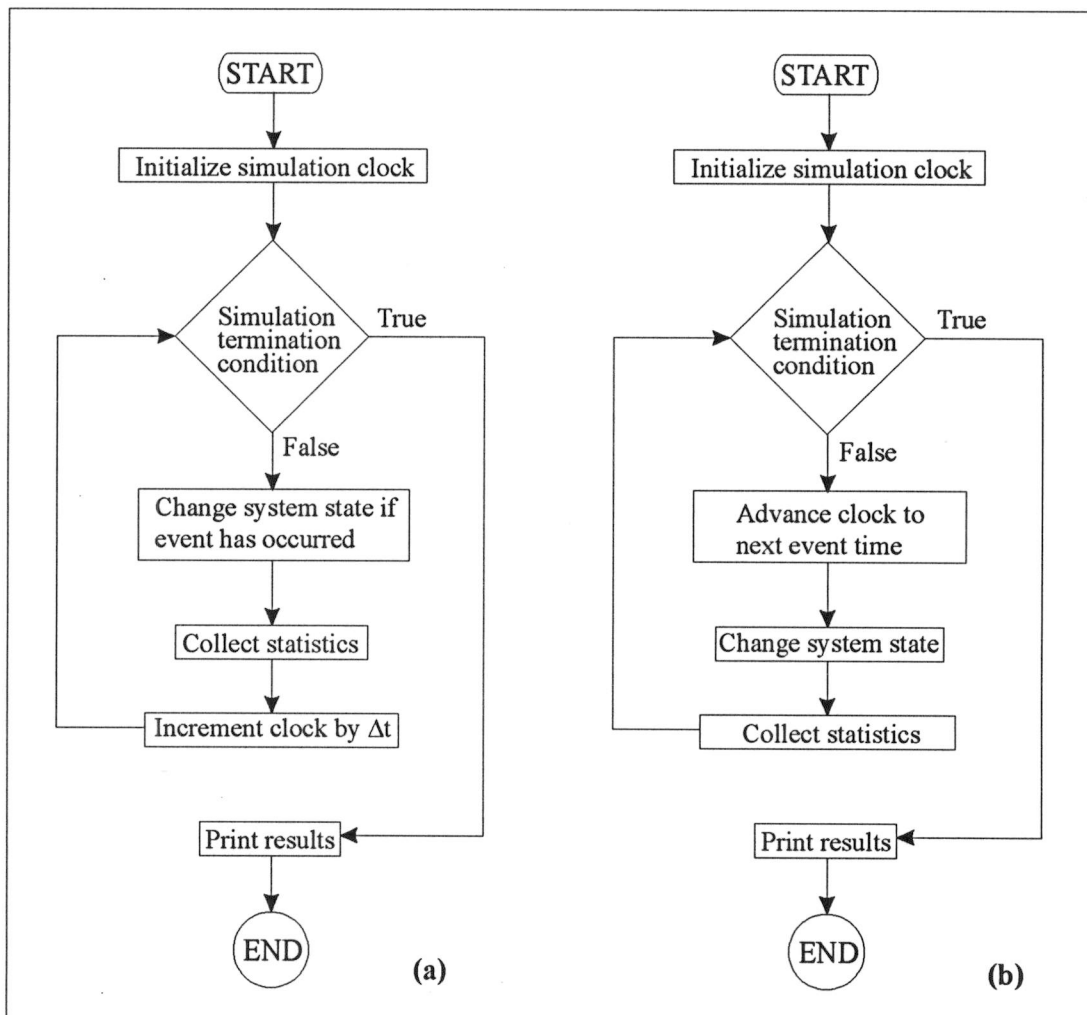


Fig. 4.1: (a) Interval-oriented simulation; (b) Event-oriented simulation

## 4.2 Simulation Program

This section describes the simulation program in terms of main program structure, event list and event types. The source code can be found in Appendix D.

### 4.2.1 Main Structure

The Program Structure Diagram (PSD) of the main program is depicted in Fig. 4.2. The simulation program simulates the working of the unslotted hybrid CDMA/ISMA protocol. First of all, the counters have to be reset to the proper values. *Clock* indicates the actual time while the simulation will last *EndTime*. While *Clock* has not reached *EndTime*, the simulation will loop. The loop is divided in four steps as shown in Fig. 4.2. In the first step, the program advances *Clock* to the next event time. This event is the first event to occur in future time. Because the event-oriented simulation is used, only events are represented explicitly in the simulation model. Depending on the event type, the program processes this event. This is the second step. The processing can generate other events or change the system state. In the third step, the generated events and the changed system states will be updated. Finally, statistics collection completes the loop. (The warm up procedure is not shown in Fig. 4.2).



Fig. 4.2: PSD of main program



### 4.2.2 Event List

From Fig. 4.2 it emerges that events and event list play an important role in the main program structure. The whole program is based on the event list. The event list consists of events linked together in an organised manner. Here, organised means in an order in which the events occur. An example of an event list is shown in Fig. 4.3. The event list consists of nodes linked together to form a chain. In the event-oriented simulation this chain symbolises the time axis. Every node represents an event and consists of an event type, the time this event occur, other information and a tail. The tail consists of a pointer pointing to the next node. The last node points nowhere. This is illustrated with the ground symbol.

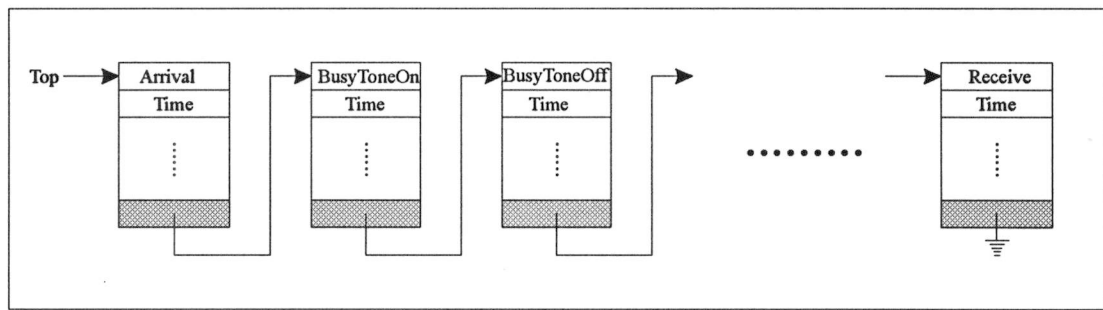


Fig. 4.3: Example of an event list

### 4.2.3 Event Types

There are five event types: *Arrival*, *Retransmission*, *BusyToneOn*, *BusyToneOff* and *Receive*. From Fig. 4.2 we see that after advancing the clock to the next event, the event will be processed depending on the event type. Those event types together with their corresponding processing are described here below.

*Arrival* - The event *Arrival* indicates the arrivals of newly arrived packets. This arrival process is modelled by the Poisson process. The inter-arrival times of the packets are negative exponentially distributed. Section 4.3 shows how the generation of this traffic stream is implemented in the simulation program.

The procedure *ProcessArrival* is illustrated in Fig. 4.4. If a new packet arrives, the next packet arrival is generated and put into the event list. This new *Arrival* event has to occur after the inter-arrival time  $T_{IA}$ , which depends on the implemented Poisson process with parameter  $\lambda$ . Because the terminals are assumed not to possess any buffers, a new packet only gets serviced if this terminal is idle. Provided that this is the case, then the state of the terminal is set to **BLOCKED**, which means that newly arrived packets will be ignored. Then, collection of the arrival data is performed. *Clock* denotes the current simulation time of the internal clock of the program.

The program makes use of a global variable *BusyTone* that can take on two values: **ON** or **OFF**. In case the *BusyTone* is **ON**, in which case the terminal is inhibited to send, then it has to wait a random delay before it can try again. Again, data collection must take place, this time it is the retransmission and delay data. If the *BusyTone* is **OFF**, then the terminal is allowed to send its data packet. First, the terminal state is altered to **TRANSMIT**. Then, after the inhibit delay fraction  $d$ , the *BusyTone* is set to **ON**. Subsequently, the event *Receive* is created. And finally, the delay data is collected for later processing.

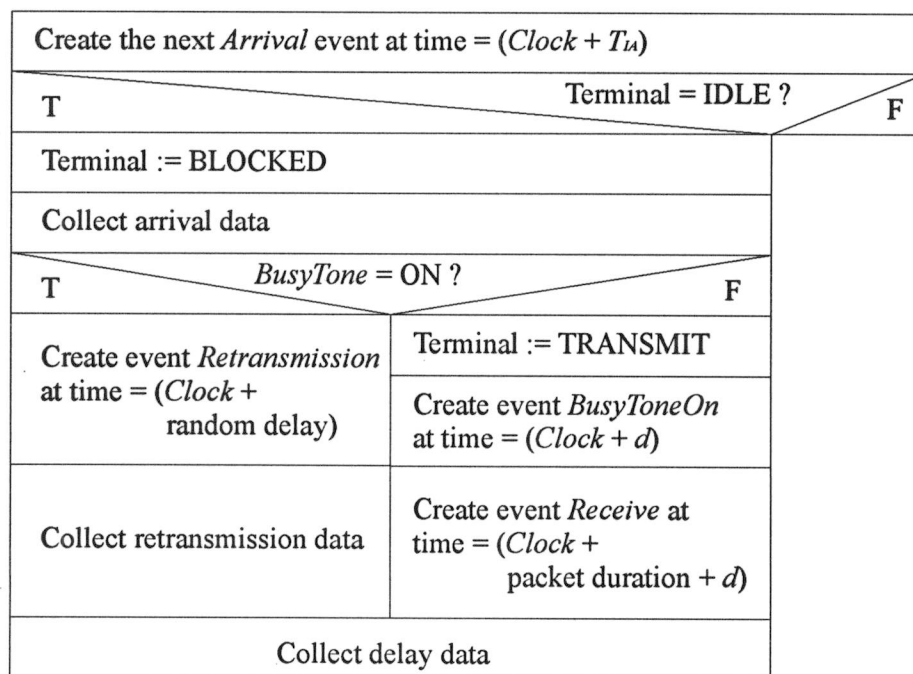


Fig. 4.4: PSD of procedure *ProcessArrival*

*Retransmission* - The event *Retransmission* indicates the arrivals of the retransmitted packets. The random delay the terminals have to wait before they are allowed to try again is also modelled by the Poisson process. The inter-arrival time of the packets is negative exponentially distributed with parameter  $\alpha$ .

Fig. 4.5 depicts the procedure *ProcessRetransmission*. The terminal senses the channel upon the arrival of a retransmitted packet. If *BusyTone* is ON, then another retransmission must be sent after a random delay. The collection of retransmission and delay data is then performed. If the *BusyTone* is OFF, then exactly the same steps as in *ProcessArrival* (after *BusyTone=ON?* is F) have to be taken.

<div style="display: flex; justify-content: space-between; padding: 5px;"> <span>T</span> <span><i>BusyTone</i> = ON ?</span> <span>F</span> </div>	
Create event <i>Retransmission</i> at time = ( <i>Clock</i> + random delay)	Terminal := TRANSMIT
	Create event <i>BusyToneOn</i> at time = ( <i>Clock</i> + <i>d</i> )
Collect retransmission data	Create event <i>Receive</i> at time = ( <i>Clock</i> + packet duration + <i>d</i> )
Collect delay data	

Fig. 4.5: PSD of procedure *ProcessRetransmission*

*BusyToneOn* and *BusyToneOff* - The events *BusyToneOn* and *BusyToneOff* take care of the inhibit mechanism of the ISMA protocol. The global variable *BusyTone* contains the right state of the *BusyTone* at any clock instant, indicating whether the channel is free or not. Fig. 4.6 and 4.7 depict the procedures *ProcessBusyToneOn* and *ProcessBusyToneOff*.

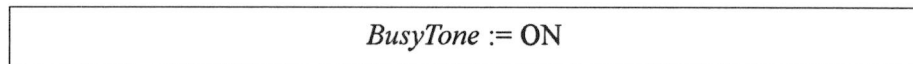


Fig. 4.6: *ProcessBusyToneOn*

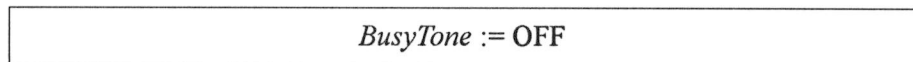


Fig. 4.7: *ProcessBusyToneOff*

*Receive* - The event *Receive* indicates the moment that the terminal knows whether the transmitted packet is received correctly or it has to be retransmitted. The procedure *ProcessReceive* is shown in Fig. 4.8. Only the first incoming packet at the base station will be serviced. If the terminal is not the first sending terminal then it has to retransmit his packet again after a random delay because its packet is ignored by the base station. This is accomplished by the event *Retransmission*. Then retransmission and delay data is collected.

T		Terminal=first sending terminal?		F
For i:=1 To all transmitting terminals			Terminal:=BLOCKED	
Terminal := INTERFERE			Create event <i>Retransmission</i> at time=( <i>Clock</i> +random delay)	
Calculate Bit Error Rate				
Calculate Packet Success Probability $P_{ps}$				
Create event <i>BusyToneOff</i> at ( $Clock_{ip}$ + packet duration + $d$ )			Collect retransmission data	
random< $P_{ps}$ ?				
T		F		
Terminal:=IDLE	Terminal:=BLOCKED			
Collect success data	Create event <i>Retransmission</i> at time=( <i>Clock</i> +random delay)			
	Collect retransmission data			
Collect delay data				

Fig. 4.8: *ProcessReceive*

If the terminal is the first sending terminal then its data packet is serviced by the base station. This does not mean that this packet is automatically received correctly by the destination. Due to the other terminals who transmit during the inhibit delay fraction ( $d$ ), data packets collide which causes interference. Because the inhibit delay fraction is only a small fraction of the packet duration and the channel is stationary (=time invariant) (see also equation 3.4), we assume that the bit error rate does not alter during the data packet. Notice that this is the worst case situation we are calculating (Fig. 4.9). If the bit error rate is known, then the packet success probability is not difficult to calculate. Fig. 4.9 also shows where the *BusyTone* has to be turned OFF. This time instant is denoted in Fig. 4.8 with

$(Clock_{ip} + \text{packet duration} + d)$ .  $Clock_{ip}$  is the time instant that the last packet is transmitted during the inhibit delay fraction. If the packet is received correctly, then the terminal state is set to IDLE to allow other newly arrived packets to get serviced. Then, the data is collected. On the other hand, if the packet is received erroneously, then the terminal state is set to BLOCKED again and a retransmission is generated. Data collection finally completes the discussion of the five event types and their processing procedures.

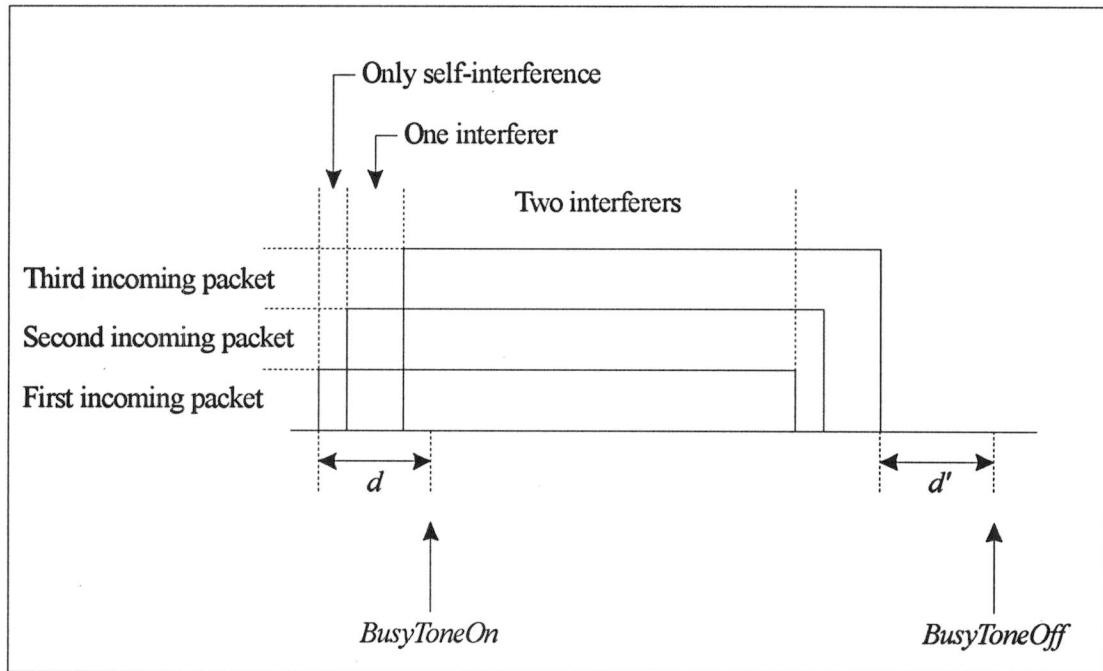


Fig. 4.9: Data packet collisions

### 4.3 The Negative Exponential Distribution

Arrivals of data packets at terminals are supposed to be generated by a Poisson process. The inter-arrival times of events of this Poisson process is negative exponentially distributed. The negative exponential distribution has a memoryless characteristic. The probability density function (PDF) and the cumulative distribution function (CDF) for the inter-arrival times are respectively given as:

$$f_T(t) = \begin{cases} \lambda \cdot e^{-\lambda t} & ; t \geq 0 \\ 0 & ; t < 0 \end{cases} \quad (4.2)$$

$$F_T(t) = \begin{cases} 1 - e^{-\lambda t} & ; t \geq 0 \\ 0 & ; t < 0 \end{cases} \quad (4.3)$$

In which  $t$  is the time it will passed till the next arrival of a packet at a terminal when on average there are  $\lambda$  arrivals per time unit [18]. The random variable  $T$  is then negative exponentially distributed with average and variance given as:

$$m_1 = \frac{1}{\lambda} \quad (4.4)$$

$$\sigma^2 = \frac{1}{\lambda^2} \quad (4.5)$$

Now we wish to generate a random variable  $T$  whose PDF and CDF are given in (4.2) and (4.3). Herefore, we make use of the so called 'Inverse Transform Method' [19]. This method functions as follows:

When the PDF  $f_X(x)$  is given, we can calculate the CDF  $F_X(x)$  by integration of  $f_X(x)$ . Otherwise, by differentiation of  $F_X(x)$ , we get  $f_X(x)$ . The value of  $F_X(x)$  varies between zero and one. So, when we wish to generate a random variable whose PDF or CDF is given, we first have to generate a random number  $U$  between zero and one with a uniform distribution. This number  $U$  functions as a value for  $F_X(x)$ . We then have:  $F_X(x) = U$ . The random variable  $X$  we want to generate follows easily:  $x = F_X^{-1}(U)$ .

The above description is of course not a mathematical proof, but serves only as background for the inverse transform method. The mathematical proof follows now.

#### 4.3.1 Proof of the Inverse Transform Method

Let  $X$  denote a random variable with CDF  $F_X$  and  $F_X^{-1}$  the inverse of  $F_X$  such that:

$$F_X^{-1}(y) = \inf\{x \mid F_X(x) \geq y\} \quad ; 0 < y < 1 \quad (4.6)$$

This means that  $F_X^{-1}(y)$  is the *infimum* or smallest value of  $X$  for which  $F_X(x) \geq y$  applies. First, we define the random variable  $Y$  as:

$$Y = F_X^{-1}(U) \quad (4.7)$$

Then,

$$\begin{aligned} Pr(Y \leq y) &= Pr[F_X^{-1}(U) \leq y] \\ &= Pr[U \leq F_X(y)] \\ &= F_X(y) = Pr(X \leq y) \end{aligned} \quad (4.8)$$

So we can conclude that  $X$  and  $Y$  have the same distribution. Therefore, applying  $F_X^{-1}$  to an uniform distribution  $U$  produces an  $x$  from  $F_X$ . This result suggests the following computing algorithm:

Let  $U$  be uniformly distributed on  $(0,1)$ . Then,

1. Generate  $U$ ;
2.  $x := F_X^{-1}(U)$ ;
3. Deliver  $x$

#### 4.3.2 Generation of the Random Numbers for a Negative Exponential Distribution

Since the CDF of the negative exponential distribution exists explicitly as

$$F_T(t) = 1 - e^{-\lambda t} \quad (4.9)$$

and its inverse is easy to calculate

$$F_T^{-1}(t) = -\frac{1}{\lambda} \cdot \ln(1-t) \quad (4.10)$$

we can use the inverse transform method to generate the random numbers for the negative exponential distribution.

**Algorithm for random numbers generation of  $F_T(t)$ :**

Let  $U$  be uniformly distributed on  $(0,1)$ , then

1. Generate  $U$ ;
2.  $t = F_T^{-1}(U) = -\frac{1}{\lambda} \cdot \ln(1-U)$
3. Deliver  $t$ .





## 5. MATHEMATICAL ANALYSIS

The performance analysis of the hybrid unslotted non-persistent CDMA/ISMA will be explained from Fig. 5.1. The time axis in Fig. 5.1 is normalised to the packet duration. If a packet arrives and the terminal senses the channel free, then the packet is sent immediately. Suppose that this time instant is  $t$ . It takes  $d$ , the inhibit delay fraction, before the busy tone is heard by all terminals. This moment occurs at the same time at all terminals because of the symmetric network configuration. Any other packet arriving between  $t$  and  $t+d$  will sense the channel free (because the busy tone has not arrived yet) and will be transmitted resulting in a conflict. Because a CDMA scheme is also used, the conflict does not necessarily result in the loss of all packets. There is a probability that the first arriving packet can still be recovered successfully. If no other terminal transmits a packet during this period  $d$ , then no conflict occurs. We model the channel between the terminals and the corresponding receivers as a multipath Rayleigh-fading channel. So, when no conflicts occur, there is still a probability that the packet can not be recovered successfully.

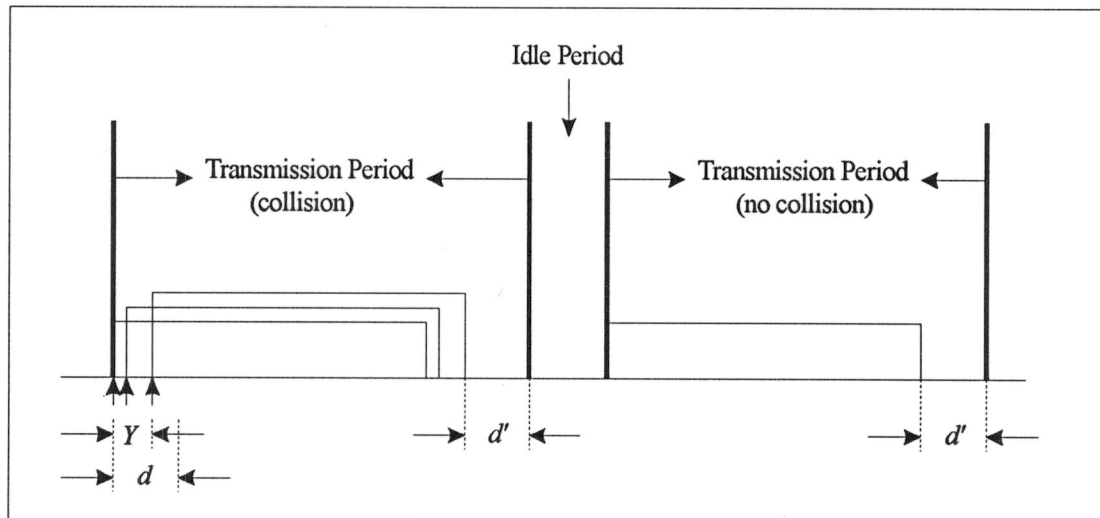


Fig. 5.1: Data cycle of the unslotted non-persistent ISMA protocol

## 5.1 Throughput

Let  $t+Y$  be the time of occurrence of the last packet arriving between  $t$  and  $t+d$ . This obviously means that  $Y$  must be between zero (the first transmitted packet is the only packet in the transmission period) and  $d$  (the end of the vulnerable period). The transmission of all packets arriving in  $(t, t+Y)$  will be completed at  $t+Y+1$ . As noted before, the channel is sensed unused only a period  $d$  later. So, any terminal becoming ready between  $t+d$  and  $t+Y+1+d$  will sense the channel busy and hence will reschedule its packet. The interval between  $t$  and  $t+Y+1+d$  is called a *transmission period (TP)*. There will be *at most* one successful transmission during a *TP*. The *idle period (IP)* is defined as the period of time between two consecutive *TP*'s. A transmission period plus the following idle period constitute a *cycle*. Let  $\overline{TP}$  be the expected duration of the transmission period,  $\overline{I}$  the expected duration of the idle period, and the average cycle time can be written as [20]:

$$\overline{t_c} = \overline{TP} + \overline{I} \quad (5.1)$$

Let  $U$  denote the time during a cycle in which the inbound channel (towards base station) is used to carry a successful packet transmission and  $\overline{U}$  the corresponding average value, then we can write the throughput as:

$$S = \frac{\overline{U}}{\overline{t_c}} \quad (5.2)$$

The expected useful time  $\overline{U}$  can be computed as follows. When a packet is successful, the channel carries useful information for a duration of  $T_{pd}$ , the packet duration. In the unsuccessful case, no useful information is carried at all or, in other words:

$$U = \begin{cases} T_{pd} & ; \text{Successful period} \\ 0 & ; \text{Unsuccessful period} \end{cases} \quad (5.3)$$

If  $P_{success}$  denotes the probability that a transmitted packet is successful then

$$\bar{U} = E[U] = T_{pd} \cdot P_{success} + 0 \cdot (1 - P_{success}) = T_{pd} \cdot P_{success} \quad (5.4)$$

As noted before, the time is normalised to  $T_{pd}$  (the packet duration), and therefore  $T_{pd}$  equals one. So, this gives us:

$$\bar{U} = P_{success} \quad (5.5)$$

To calculate the average idle period we make the assumption that the total rate at which users schedule new and retransmitted packets forms a Poisson process with parameter  $G$ . So, new and rescheduled packets arrive at a rate of  $G$  packets per unit time. Not all arrivals do result in a transmission. If a packet arrives and the terminal finds the channel in a busy state then the packet is rescheduled for transmission at a later time instant. In literature,  $G$  is also called the offered channel traffic.

If the packet arrival times are independent and exponentially distributed with a mean arrival rate of  $G$  packets per second, the probability of  $k$  arrivals in an interval of duration  $t$  is then a Poisson process given by  $P_k(t)$  [18], where

$$P_k(t) = \frac{(Gt)^k \cdot e^{-Gt}}{k!} \quad (5.6)$$

For our work, the time is normalised to the packet duration ( $T_{pd}$ ). This means that  $G$  denotes the number of packet arrivals per *packet duration* and the time unit is also measured in *packet duration* instead of seconds. The probability of the idle time being greater than some value  $t$  is the probability that no packets are scheduled within a time interval of duration  $t$  and with the assumed Poisson packet scheduling process this probability becomes:

$$P(I \geq t) = P(\text{no packet arrival in interval of duration } t) = e^{-Gt} \quad (5.7)$$

Therefore the average value of  $I$  can be expressed as:

$$\bar{I} = \frac{1}{G} \quad (5.8)$$

The average duration of a transmission period equals to:

$$\overline{TP} = 1 + \overline{Y} + d \quad (5.9)$$

where  $\overline{Y}$  is the expected value of  $Y$ . Since  $Y$  denotes the time at which the last interfering packet is scheduled, the probability of  $Y$  being smaller than some time  $y$  is the probability that no other packets (either new or retransmissions) are scheduled for transmission in an interval of duration  $d-y$ . With Poisson arrivals, the distribution for  $Y$  is:

$$\begin{aligned} F_Y(y) \triangleq \Pr\{Y \leq y\} &= \Pr\{\text{no arrival occurs in an interval of length } d-y\} \\ &= e^{-G(d-y)} \end{aligned} \quad (5.10)$$

The average of  $Y$  is therefore given by:

$$\overline{Y} = d - \frac{1}{G} (1 - e^{-Gd}) \quad (5.11)$$

Combining the equations (5.1), (5.8), (5.9) and (5.11), we get:

$$\overline{t}_c = \overline{TP} + \overline{I} = 1 + \overline{Y} + d + \overline{I} = 1 + \left[ d - \frac{1}{G} (1 - e^{-Gd}) \right] + d + \frac{1}{G} \quad (5.12)$$

$$S = \frac{\overline{U}}{\overline{t}_c} = \frac{P_{\text{success}}}{1 + \left[ d - \frac{1}{G} (1 - e^{-Gd}) \right] + d + \frac{1}{G}} = \frac{G \cdot P_{\text{success}}}{G \cdot (1 + 2d) + e^{-Gd}} \quad (5.13)$$

The term  $P_{\text{success}}$  in (5.7) is the only term that needs to be specified. Herefore we distinguish between four types of transmissions during a transmission period:

1. Successful transmission without conflict
2. Unsuccessful transmission without conflict
3. Successful transmission with conflict

#### 4. Unsuccessful transmission with conflict

If the channel is used for the delivering of a successful packet during a  $TP$  then this is denoted by a successful transmission (situation 1 or 3). The difference between a transmission with or without conflict can be found in the number of packets transmissions during a  $TP$ . In case there is more than one packet transmission during a  $TP$ , we speak of a transmission with conflict. If there is only one packet transmission then the transmission is conflict-free. Keeping this in mind, we can divide  $P_{success}$  up into two parts:

$$P_{success} = P_{success|noconflict} + P_{success|conflict} \quad (5.14)$$

In which  $P_{success|noconflict}$  corresponds to situation 1 and  $P_{success|conflict}$  to situation 3.

$$P_{success|noconflict} = e^{-Gd} P_{ps} \quad (5.15)$$

This is equal to the probability that no terminal transmits during the inhibit delay fraction  $d$  multiplied by the packet success probability  $P_{ps}$ . The multiplication with  $P_{ps}$  is necessary because of the assumption of the Rayleigh fading channel.

$$P_{ps} = \left(1 - P_{be|x=0}\right)^{L_p} \quad (5.16)$$

$P_{be|x=0}$  is the bit error probability in Rayleigh fading channel using the CDMA scheme and is caused by self interference due to multipath.  $L_p$  denotes the packet length; in our model the packet is 64 bits long. The calculation of  $P_{success|noconflict}$  is not very complicated. However, this is not the case for  $P_{success|conflict}$  which is the probability that the first packet has arrived successfully given a collision (between two or more packets).

$$P_{success|conflict} = \sum_{x=1}^{K-1} P_x(d) \cdot P\{\text{first packet OK}|x\} \quad (5.17)$$

in which:

$K$  is the number of active users in the system

$P_x(d)$  is the probability of  $x$  arrivals during  $d$  for a Poisson distribution

$x$  denotes the number of arrivals during  $d$

Terminals within one group are supposed to be identical in every aspect. They have the same arrival rate, the same code and the same distance to their own transceiver. Notice that the distance between a terminal and the base station does not play a role. This can be explained as follows. If a terminal is allowed to transmit its data packet then it will do so using radio communication to deliver its packet to its transceiver. The transceiver on its turn forwards the received packet to the base station by wire. The base station only process the first incoming packet arriving by wire (see also Fig. 5.2). All other packets will be ignored by the base station.

Suppose that the network configuration in Fig. 5.2 is deployed.  $T_i$  denotes terminal number  $i$  and  $TR_j$  the transceiver number  $j$ . The data packet transmitted by  $T_i$  is denoted by  $p_i$ . For example, we adopt the following scenario.  $T_1$  starts sending its data packet  $p_1$  at time instant  $t$  and  $T_7$  a fraction later, at  $(t + \Delta t)$  in which  $\Delta t \leq d$ . The data packet  $p_1$  arrives  $\Delta t$  earlier at its transceiver  $TR_1$  than its competitor  $p_7$  at  $TR_2$ . Consequently,  $TR_1$  forwards its received packet  $\Delta t$  earlier to the base station than  $TR_2$ . The base station notices that  $TR_1$  is the first active receiver and processes this packet.  $TR_2$ ,  $TR_3$  and  $TR_4$  are all ignored. Even though the data packet  $p_7$  is ignored by the base station, this packet causes interference for  $p_1$ . The radio transmitted packet  $p_7$  not only arrives at  $TR_2$  (its own receiver), but also at other receivers ( $TR_1$ ,  $TR_3$  and  $TR_4$ ). Consequently  $TR_1$  not only receives  $p_1$ , but also the interfering packet  $p_7$ . This interference is less severe than from the interference caused by  $T_2$ ,  $T_3$  or  $T_4$ . The explanation herefore is twofold. First,  $T_2$ ,  $T_3$  and  $T_4$  all have the same code as  $T_1$  and second, the received interference power (at  $TR_1$ ) of  $T_2$ ,  $T_3$  or  $T_4$  are higher than  $T_7$  because of the shorter distance to  $TR_1$ .

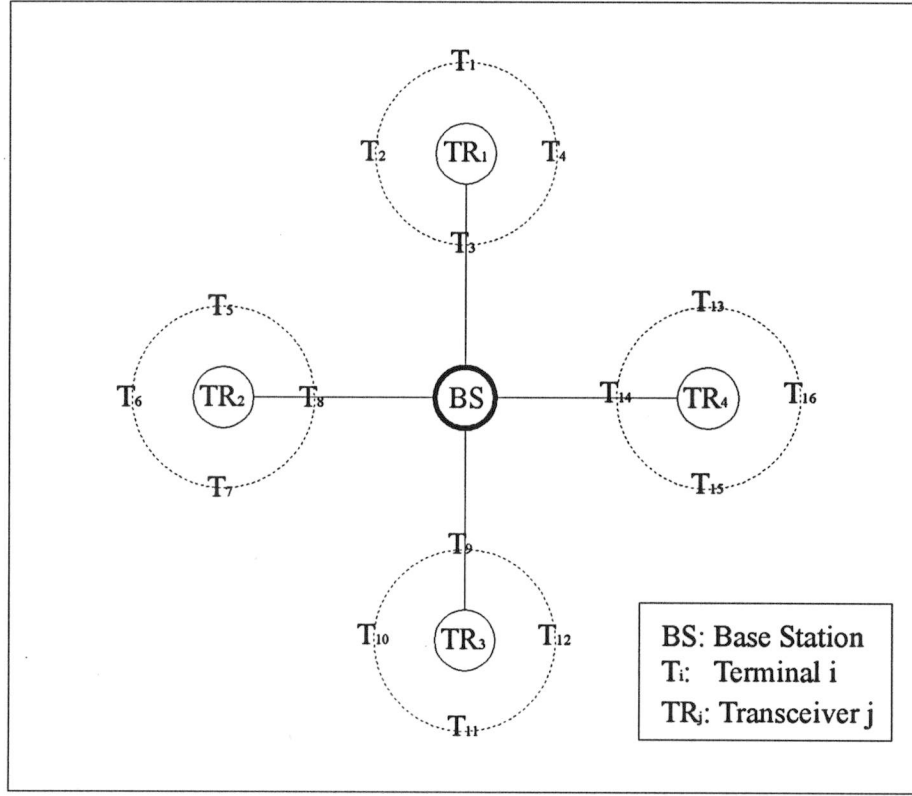


Fig. 5.2: Example of a 16-terminal network configuration

Without loss of generality, suppose that  $T_1$  is the reference terminal. This terminal sends its data packet at an idle period first. Other terminals sending within the inhibit delay fraction interfere with the reception of the data packet  $p_1$ . The bit error rate depends on the configuration of transmitting terminals.  $P\{\text{first packet OK}|x\}$  can then be evaluated as:

$$P\{\text{first packet OK}|x\} = E\left[1 - P_{be|\bar{x}}\right]^{L_p} \quad (5.18)$$

$E[.]$  denotes the expectation value, and  $\bar{x}$  a certain configuration of  $x$ . The total number of combinations for  $\bar{x}$  equals  $\binom{K-1}{x}$ , in which  $\binom{K-1}{x} = \frac{(K-1)!}{x!(K-1-x)!}$ .

Because we assume an uniform distribution for all possible configurations we can write ( $K$  is the total number of active users in the system):



$$E[1 - P_{be|\bar{x}}]^{L_p} = \frac{1}{\binom{K-1}{x}} \cdot \sum_{\bar{x}} (1 - P_{be|\bar{x}})^{L_p} \quad (5.19)$$

$$P_{success|conflict} = \sum_{x=1}^{K-1} P_x(d) \cdot \frac{1}{\binom{K-1}{x}} \cdot \sum_{\bar{x}} (1 - P_{be|\bar{x}})^{L_p} \quad (5.20)$$

$$P_{success} = e^{-Gd} (1 - P_{be|x=0})^{L_p} + \sum_{x=1}^{K-1} P_x(d) \cdot \frac{1}{\binom{K-1}{x}} \cdot \sum_{\bar{x}} (1 - P_{be|\bar{x}})^{L_p} \quad (5.21)$$

Combining the equations (5.13) and (5.21) finally gives:

$$S = \frac{G \left[ e^{-Gd} (1 - P_{be|x=0})^{L_p} + \sum_{x=1}^{K-1} P_x(d) \cdot \frac{1}{\binom{K-1}{x}} \cdot \sum_{\bar{x}} (1 - P_{be|\bar{x}})^{L_p} \right]}{G(1+2d) + e^{-Gd}} \quad (5.22)$$

For large networks ( $K \geq 32$ ), the calculation of the second term in the numerator can take long. If we take a closer look at this term we can distinguish two summations. For example, we take a 32 terminal network, and this simplified term looks like:

$$\sum_{x=1}^{31} \sum_{\binom{31}{x}} \dots \quad (5.23)$$

We see that the value of  $x$  determines the duration of the calculations. Obviously, it is not possible to derive equation (5.23) for  $x = 1$  up to 31. The total number of combinations will be over 4 billions! Fortunately, this is also not necessary. We can limit  $x$  to a certain extent. To explain this we have to return to equation (5.17). From (5.17) we see that  $P_{success/conflict}$  depends on the multiplication of  $P_x(d)$  with  $P\{\text{first packet OK} | x\}$ . Fig. 5.3 and 5.4 depict

$P_x(d)$  versus  $G$  with  $x$  and  $d$  as parameter respectively.  $P_x(d)$  is derived from formula (5.6). For high values of  $x$ ,  $P_x(d)$  drops quickly. Because high values of  $x$  is identical to a high number of interferes,  $P\{\text{first packet OK}|x\}$  also drops quickly for increasing values of  $x$ . Therefore, the contribution in formula (5.17) for high values of  $x$  is negligible. Consequently, we can limit  $x$  to a certain value. There is a trade off between the calculation time and the accuracy of the calculation.

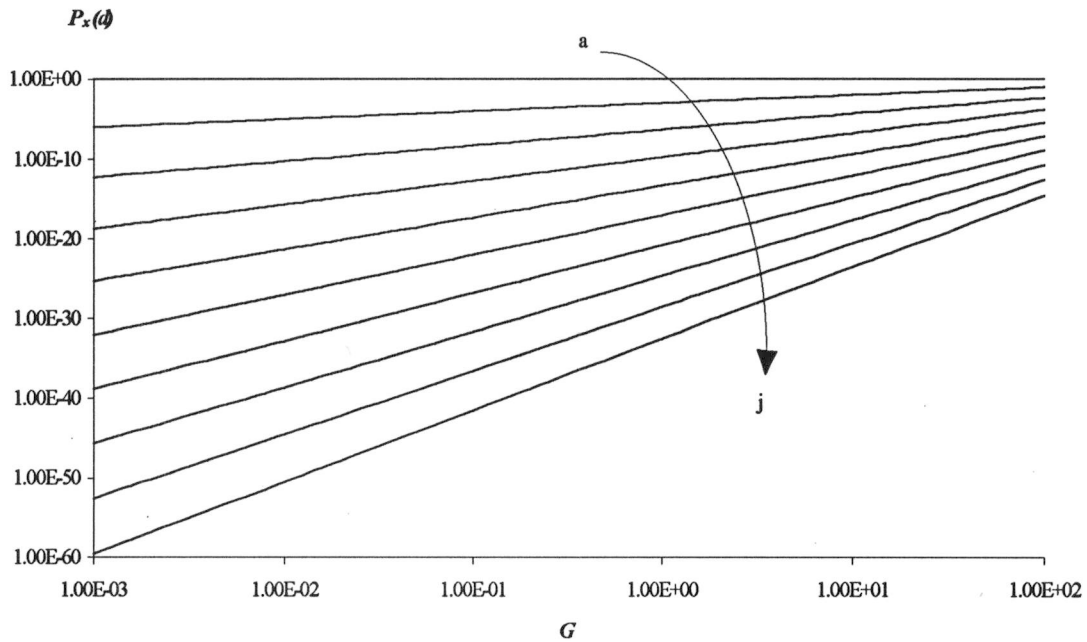


Fig. 5.3:  $P_x(d)$  versus  $G$  with  $x$  as parameter; a..j:  $x=0, 1, 2, \dots, 8, 9$ ;  $d = 0.001$

We adopt the following strategy for the derivation of the throughput for the hybrid protocol. The throughput will be derived by formula (5.22) for increasing limit value of  $x$ . After a certain value, the throughput with  $x$  do not differ much from the throughput with  $(x-1)$  and the calculation time will last too long. Depending on the achieved results we can choose a reasonable and considered limit value of  $x$ . Section 5.3 deals with this aspect in more detail.

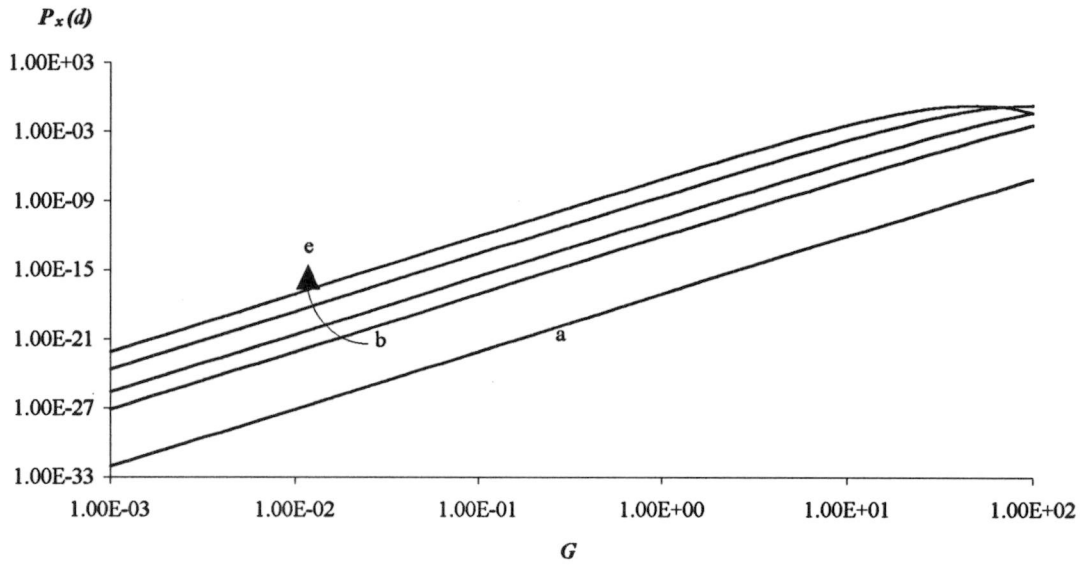


Fig. 5.4:  $P_x(d)$  versus  $G$  with  $d$  as parameter; a..e:  $d=0.001, 0.01, 0.02, 0.05$  and  $0.1$ ;  $x = 5$

## 5.2 Delay

The average packet delay is defined as the duration between the transmission of the first bit till the correct reception of the last bit. For the calculation of the packet delay of the unslotted hybrid CDMA/ISMA protocol, we use the block diagram depicted in Fig. 5.5.

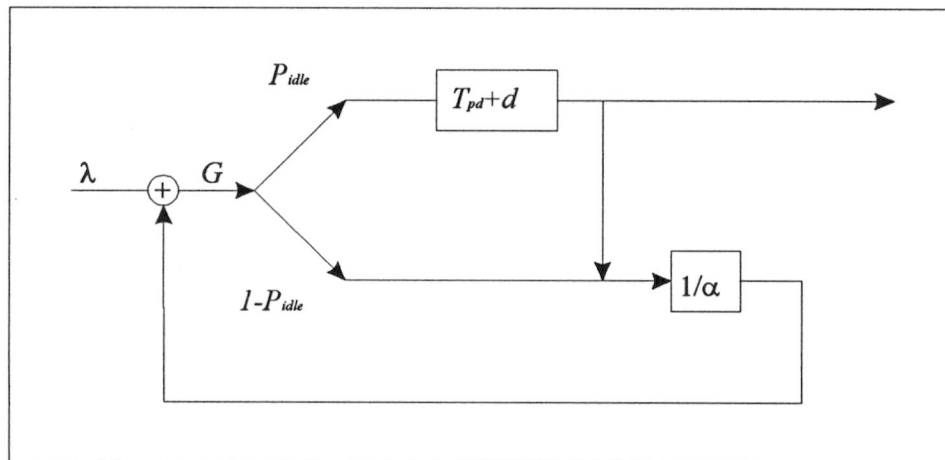


Fig. 5.5: Delay in the unslotted non-persistent hybrid CDMA/ISMA protocol

If a new packet arrives, the terminal immediately senses the channel to decide if it can transmit the packet. In case the channel is sensed busy (probability  $1-P_{idle}$ ), the packet is rescheduled for a later time instant. This random delay process is assumed to be negative exponentially distributed with parameter  $\alpha$ . The average random delay time is then  $1/\alpha$ . The average number of times the packet has to suffer this delay is  $G(1-P_{idle})/S$  (see Fig. 5.5).  $S$  is the throughput as given by (5.22).

On the other hand, if the channel is sensed idle (probability  $P_{idle}$ ), the packet will be transmitted immediately. This transmission will take a packet duration  $T_{pd}$  plus the inhibit delay fraction  $d$  before the knowledge about the outcome of the transmission is available (see also Fig. 5.5). So, the total delay until the terminal knows whether the transmission was successful or not is  $(T_{pd} + d)$ . The transmission of a data packet does not necessarily result in a successful transmission; there are two possibilities: a successful or a failed transmission. If the transmission was successful, the packet leaves the system. The delay this packet suffers is  $(T_{pd} + d)$ . If the transmission was a failure, the packet again has to wait a random delay. In this case, the average delay is  $(T_{pd} + d + \frac{1}{\alpha})$  and the average number of schedulings is  $(\frac{GP_{idle}}{S} - 1)$ .

Combining the above results, the average delay is finally given by equation (5.24).

$$D = \left( \frac{GP_{idle}}{S} - 1 \right) \left[ T_{pd} + d + \frac{1}{\alpha} \right] + \frac{G(1-P_{idle})}{S} \cdot \frac{1}{\alpha} + (T_{pd} + d) \quad (5.24)$$

Again, because  $T_{pd}$  equals to 1, (5.24) can be rewritten as:

$$D = \left( \frac{GP_{idle}}{S} - 1 \right) \left[ 1 + d + \frac{1}{\alpha} \right] + \frac{G(1-P_{idle})}{S} \cdot \frac{1}{\alpha} + (1 + d) \quad (5.25)$$

$S$  can be found in formula (5.22) and  $P_{idle}$  still has to be calculated. The average probability of sensing the channel idle is the average probability that the channel is idle and this probability equals to the average idle time ( $\bar{I}$ ) divided by the average cycle length ( $\bar{t}_c$ ).

$$P_{idle} = \frac{\bar{I}}{\bar{t}_c} = \frac{\frac{1}{G}}{\frac{1}{G} + \left\{ 1 + d - \frac{1}{G}(1 - e^{-Gd}) + d \right\}} = \frac{1}{G(1 + 2d) + e^{-Gd}} \quad (5.26)$$

### 5.3 Practical Implementation Aspect

Formula (5.22) can not be calculated exactly if we do not possess strong computing power. However, formula (5.22) allows us to approach the exact value very accurately in case we only have limited computing power.

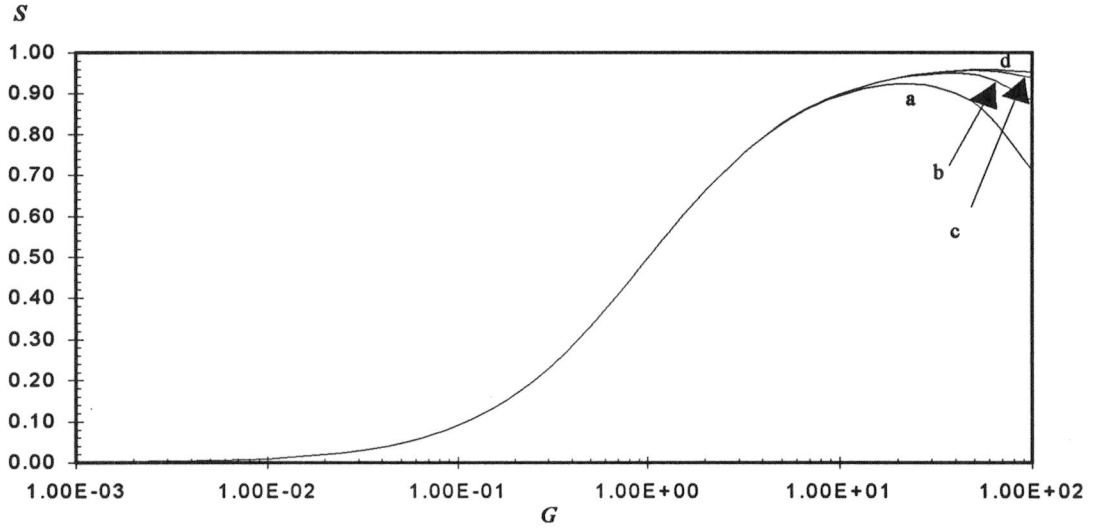


Fig. 5.6: Throughput versus offered traffic; a..d:  $x=1.4$ ;  $\alpha=0.1$ ;  $d=0.01$ ; code length=31; SNR=20 dB; number of codes=8; number of terminals=32

The influence of  $x$  in formula (5.22) on the results is depicted in Fig. 5.6 and 5.7. As explained from the previous section,  $x$  denotes the number of arrivals during the inhibit delay

fraction  $d$  and determines for a great deal the calculation duration. From Fig. 5.6 and 5.7 in which the throughput and delay are depicted respectively, we can see that the protocol performance converges to a certain final level. (The delay time is normalised to the packet duration  $T_{pd}$ ). The difference between  $x=3$  and  $x=4$  is hardly noticeable. We adopt  $x=4$  as a reasonable value for our further analysis. However, for higher values of the offered traffic  $G$ , this difference increases. This can be explained by the fact that if the offered traffic increases, more packets arrive during the inhibit delay fraction.

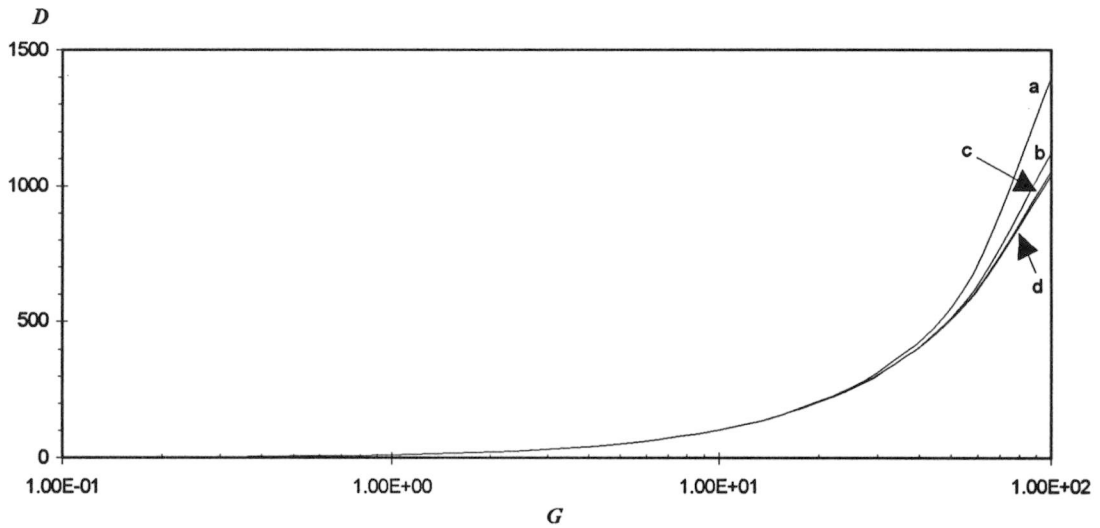


Fig. 5.7: Delay versus offered traffic; a..d:  $x=1..4$ ;  $\alpha=0.1$ ;  $d=0.01$ ; code length=31;  
SNR=20dB; number of codes=8; number of terminals=32

The mathematical analysis in this chapter is nice in the sense that it allows us to make a trade off between accuracy and computing time. If we do not have strong computing power or computing time, we are also able to obtain the performance result of the hybrid protocol. However, we have to live with a less accurate result. On the other hand, if accuracy is necessary, then we have to pay for strong computing power.



## 6. RESULTS

This chapter presents the results obtained from simulation and mathematical analysis as explained in the previous chapters. The protocol performance has been measured in terms of throughput  $S$  and delay  $D$  in relationship with the offered traffic  $G$ . The throughput is a measure of the efficiency of the protocol and is defined as the fraction of time in which correct data packets are received. The delay is the time between the arrival of the first bit of a data packet at a terminal and its arrival of the last bit at the destination. The delay is measured in terms of the packet duration.

In section 1, we first compare the performance between the slotted and unslotted protocol using simulation. Then, in section 2, we discuss the comparison between the simulation and mathematical model for the unslotted protocol. The conditions under which the results are valid can be found in section 2.6. If a parameter is changed, then it will be stated explicitly.

### 6.1 Comparison between Slotted and Unslotted Protocol

The performance shown in this section is generated by simulation. Fig. 6.1a and b compare the performance of the slotted and unslotted hybrid protocol using computer simulation. In these figures, we also show the influence of  $P_{tr}$  on the performance of the slotted protocol.  $P_{tr}$  is one of the most important parameters for the slotted protocol and is described in paragraph 2.4.1 as the probability that the packet will actually be transmitted in case the channel is sensed *free*. As we can observe, for increasing value of  $P_{tr}$ , the throughput improves. The declaration for this observation can be found by the use of direct sequence CDMA. If only pure ISMA is used, than it is to be expected that for high traffic, the increase in  $P_{tr}$  leads to a degradation of the protocol's performance. High values of  $P_{tr}$  means a high number of simultaneously transmitted data packets. Collision between those packets results in corruption of all data packets. However, because CDMA is also used, there is still a probability that the first incoming packet can be received correctly.



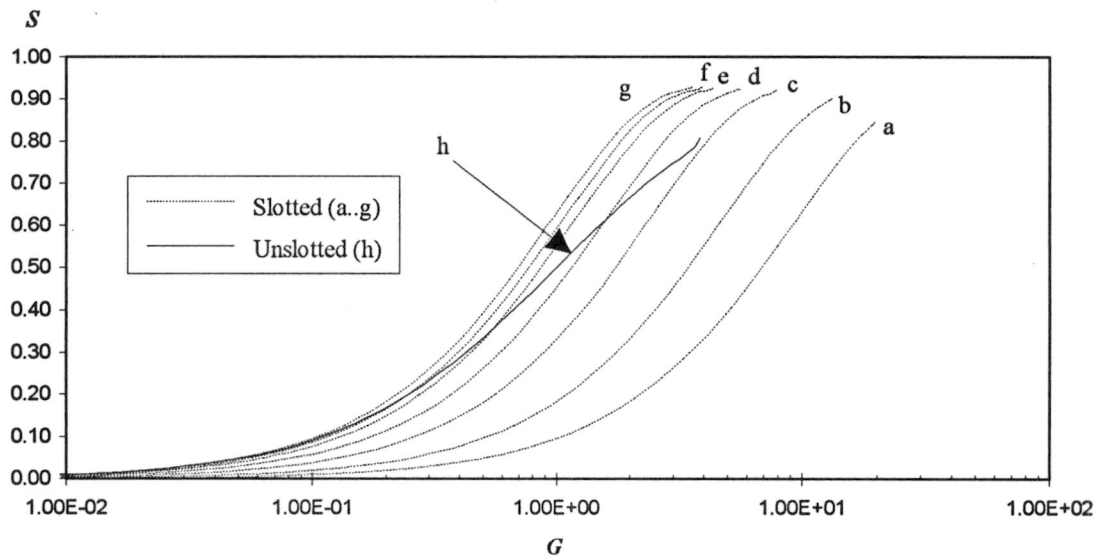


Fig. 6.1a: Throughput versus offered traffic using simulation of slotted (a..g:  $P_{tr}=0.1, 0.2, 0.4, 0.6, 0.8, 0.9$  and  $1$ ) and unslotted (h:  $d=0.01$ ) hybrid protocol; 8 codes

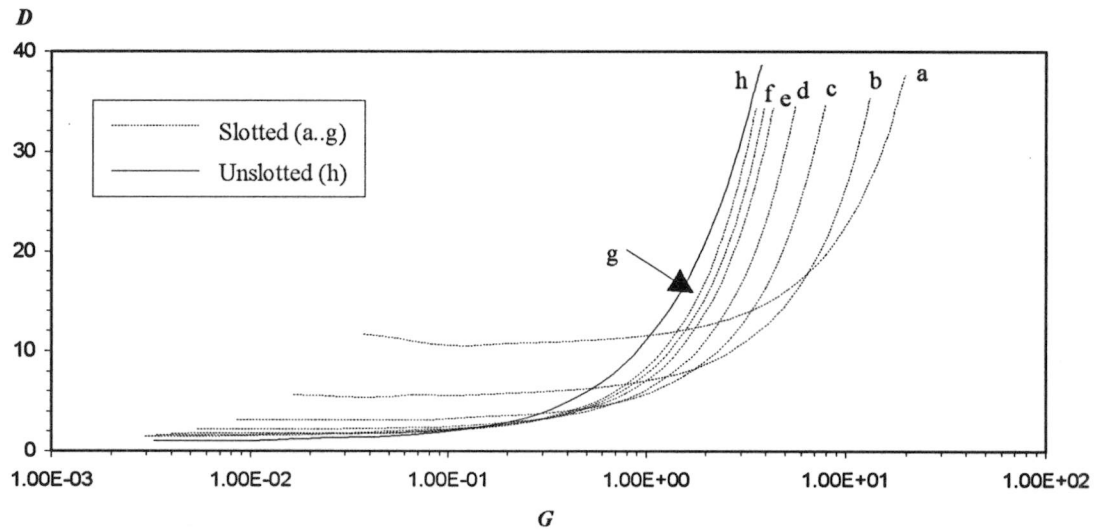


Fig. 6.1b: Delay versus offered traffic using simulation of slotted (a..g:  $P_{tr}=0.1, 0.2, 0.4, 0.6, 0.8, 0.9$  and  $1$ ) and unslotted (h:  $d=0.01$ ) hybrid protocol; 8 codes

For the delay, there is one interesting observation we can make. If we look at low offered traffic, then the delay for low  $P_{tr}$  is high. For example, the average delay for  $P_{tr}=0.1$  is about 10 packet durations. This is not surprising because the probability of actually transmitting a

packet if the channel is sensed idle equals to 0.1. So, on average, a packet has to wait 10 packet durations before it can be transmitted. On the other hand, if we look at high offered traffic, the delay is lower for low values of  $P_{tr}$ . For example, take a look at the delay curves of  $P_{tr}=0.1$  and  $0.2$  (curves a and b of Fig. 6.1b). For an offered traffic below 8 packets/(packet duration), the delay for  $P_{tr}=0.1$  is higher than for  $P_{tr}=0.2$ . For an offered traffic beyond 8 packets/(packet duration) it is the other way round. So, if we have to choose a slotted scheme, then the region of the offered traffic in which we are, decides of value for the parameter  $P_{tr}$ . If the offered traffic is low, than a slotted ISMA protocol with a high value of  $P_{tr}$  is better.

The throughput of the unslotted protocol is better than the throughput of the slotted protocol for low values of the offered traffic. For increasing values of the offered traffic, the throughput of the unslotted protocol starts to drop below the throughput of the slotted protocol. This is because for high offered traffic, the slotted protocol organises its data stream better than the unslotted protocol. Packets are only allowed to be transmitted at the start of a time slot. Furthermore, the slotted protocol with low values of  $P_{tr}$  handles high offered traffic quite well, because it limits of the contention in the channel more profoundly. For low traffic, the delay of the unslotted protocol is, as expected, better than the slotted protocol. For high traffic, the situation is the other way round.

Fig. 6.2 shows the influence of code sharing on the performance of the slotted and unslotted protocol.  $P_{tr}$  plays an important role in the slotted scheme. This is the same for  $d$  with respect to the unslotted scheme. The slotted protocol is simulated for  $P_{tr}=0.9$ . As already noted in section 2.6, the inhibit delay fraction  $d$  equals to 0.01 for the unslotted protocol.

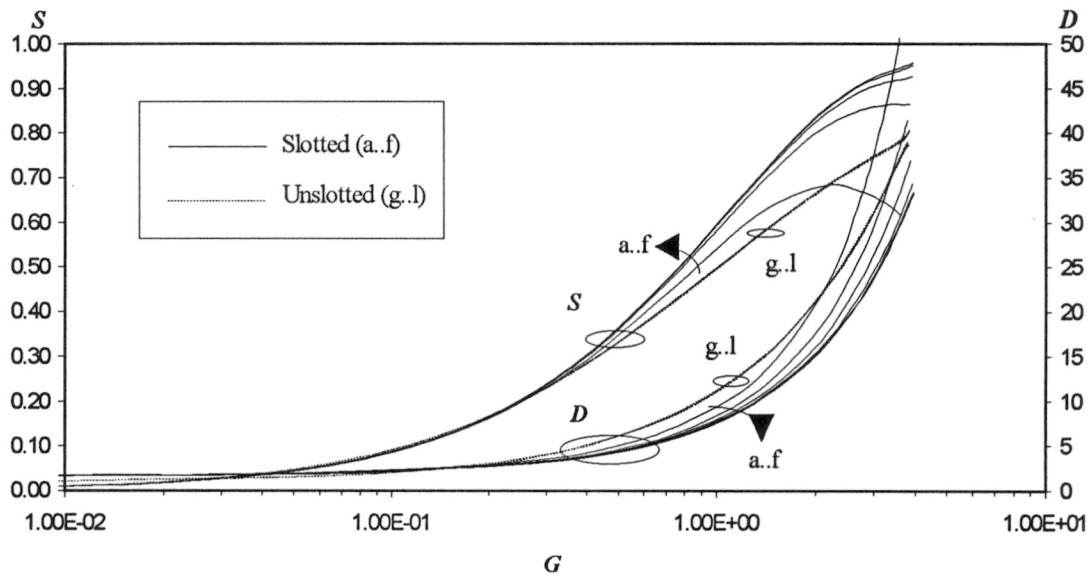


Fig. 6.2: Throughput and delay versus offered traffic using simulation for the slotted ( $P_r=0.9$ ) and unslotted ( $d=0.01$ ) protocol, varying the number of codes a..f: g..l: number of codes: 1, 2, 4, 8, 16 and 32.

## 6.2 Comparison between Simulation and Mathematical Analysis

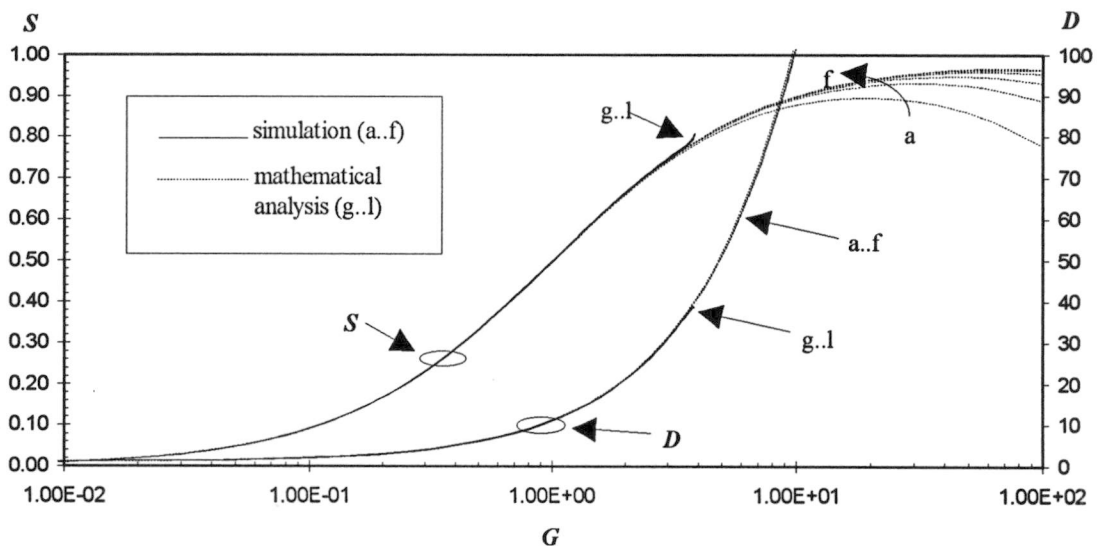


Fig. 6.3: Comparison between simulation and mathematical analysis for the unslotted protocol. a..f, g..l: 1, 2, 4, 8, 16, 32 codes

The comparison between the simulation and mathematical model for the unslotted protocol is given in Fig. 6.3. The performance of simulation does not exceed for  $G \approx 4$  packets/(packet duration). This is because in case of simulation, the arrival rate at terminals is the input parameter and  $G$  is one of the output parameters. The results due to simulation and mathematical model are in good agreement. This graph also shows the effect of code sharing. For a 32-terminal network, it is sufficient to use only four codes. If we take a closer look at the simulation results, it seems like code sharing has no influence at all on the performance. Fig. 6.4a and b show the influence of the number of codes used on the performance of the unslotted protocol for poor parameters. In contrast to Fig. 6.3 where good parameters are chosen, Fig. 6.4a and b show that the number of codes used do influence protocol performance. The more codes, the better the protocol performs. However, code sharing is a very good option. The differences between 8 and 32 codes is much lower than the difference between 1 and 32 codes.

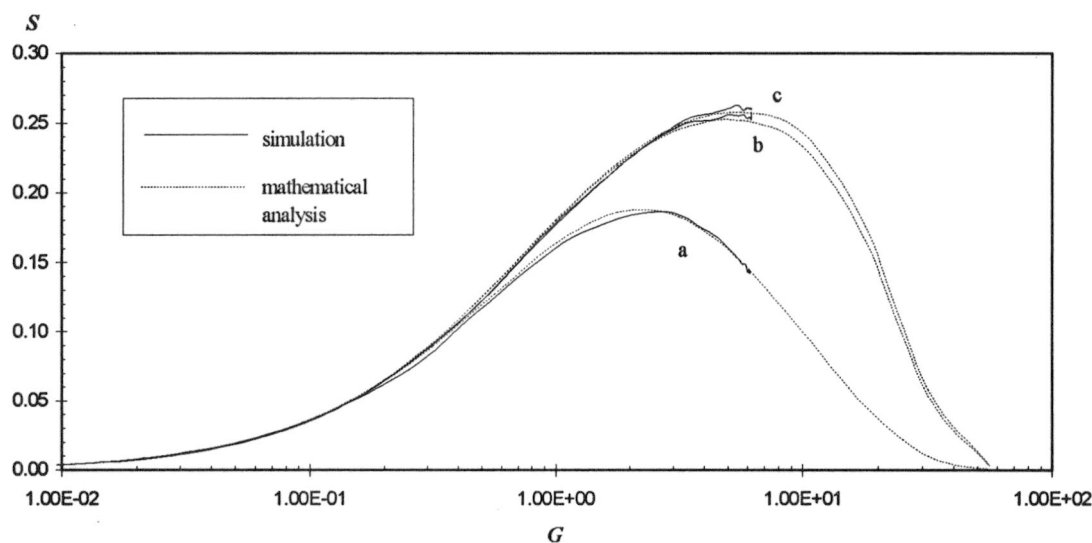


Fig. 6.4a: Throughput comparison between simulation and mathematical analysis for the unslotted protocol and varying the number of codes used; a..c: number of codes=1, 8 and 32;

$d=0.2$ ;  $\alpha=0.2$ ; SNR=6 dB; code length = 31

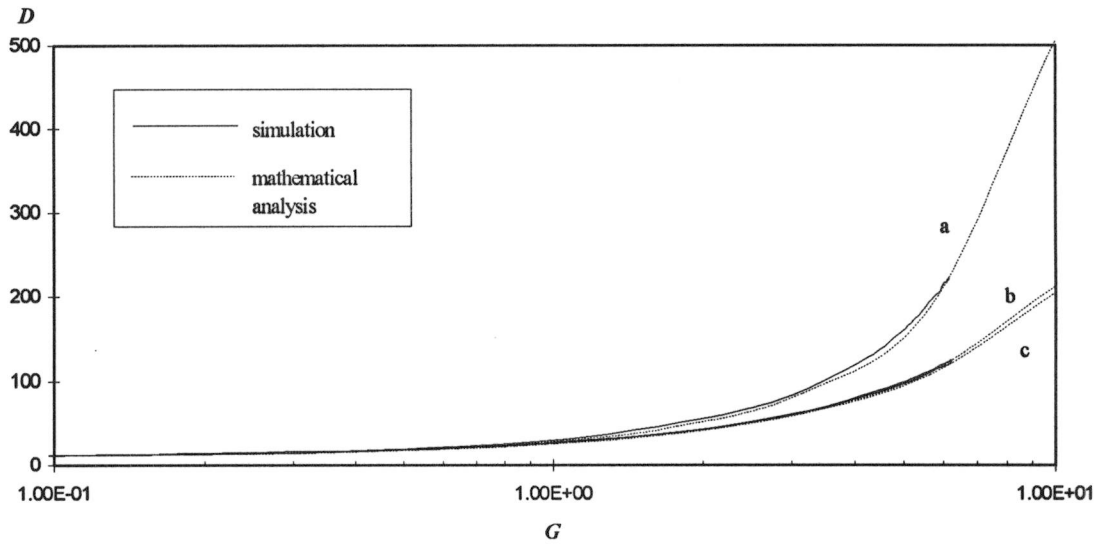


Fig. 6.4b: Delay comparison between simulation and mathematical analysis for the unslotted protocol and varying the number of codes used; a..c: number of codes=1, 8 and 32;  $d=0.2$ ;  $\alpha=0.2$ ; SNR=6 dB; code length = 31

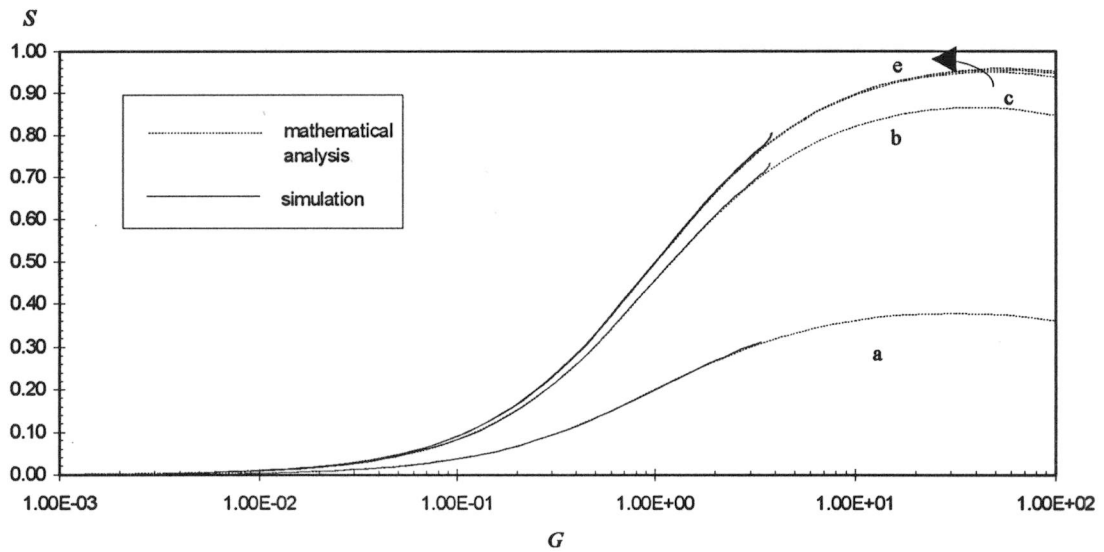


Fig. 6.5a: Throughput comparison between simulation and mathematical analysis for the unslotted protocol and varying SNR; a..e: SNR=6, 8, 10, 12 and 20 dB; number of codes=8

Fig. 6.5a and b show the effect of the signal to noise ratio (SNR) on the performance of the unslotted protocol. Of course, the throughput and delay improve for increasing SNR.

However, Fig. 6.5a and b also show that a SNR of 10 dB suffice. For values of SNR higher than 10 dB, the protocol performance also improves, but only for high values of the offered traffic and the improvement is only additional. Again, simulation and mathematical analysis agree very well with each other.

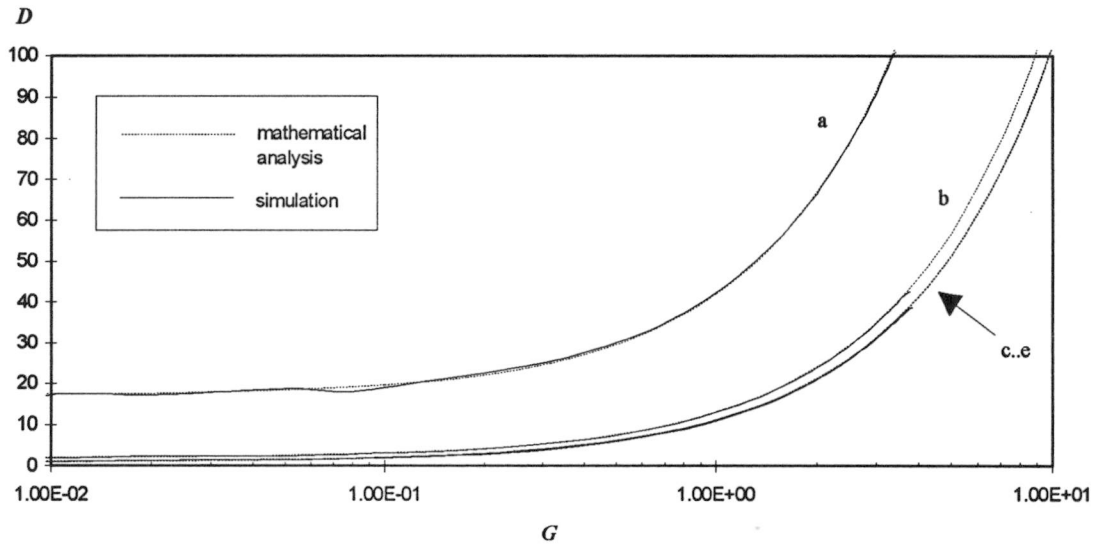


Fig. 6.5b: Delay comparison between simulation and mathematical analysis for the unslotted protocol and varying SNR; a..e: SNR=6, 8, 10, 12 and 20 dB; number of codes=8;

Fig. 6.6a and b show the effect of the inhibit delay fraction  $d$  on the performance of the unslotted protocol. For high values of  $d$ , the performance degrades tremendously. This is due to the increase of the number of data packets colliding during this period. This is why the ISMA protocol is not useful for environment in which the propagation delay is high.

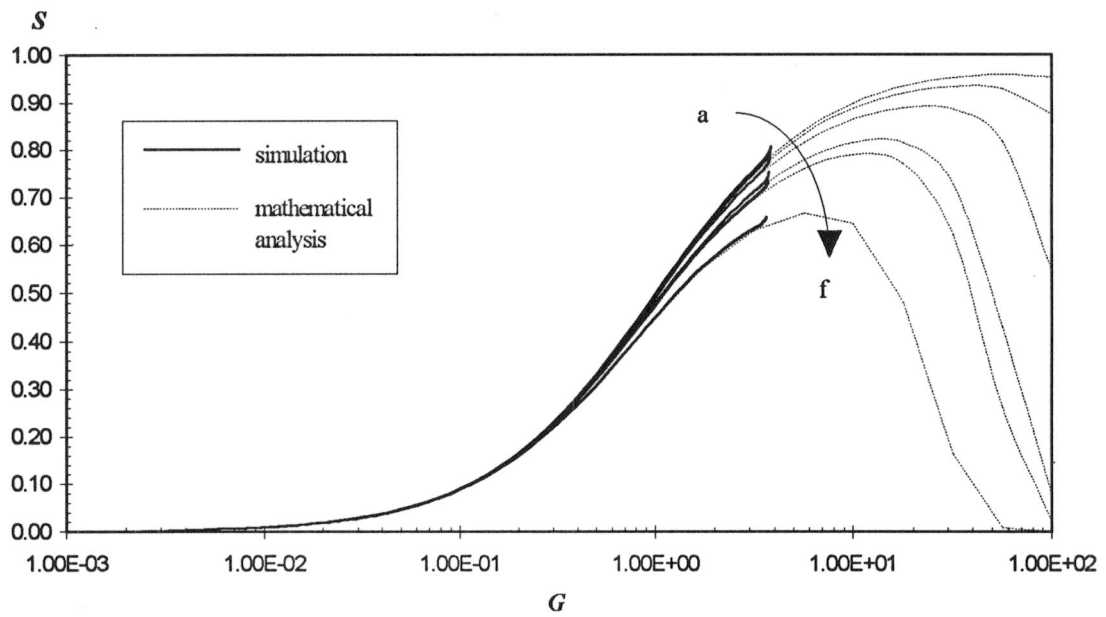


Fig. 6.6a: Throughput versus offered traffic using simulation and mathematical analysis of unslotted protocol, varying the inhibit delay fraction  $d$ . a..f:  $d=0.01, 0.02, 0.04, 0.08, 0.1$  and  $0.2$ ; 8 codes

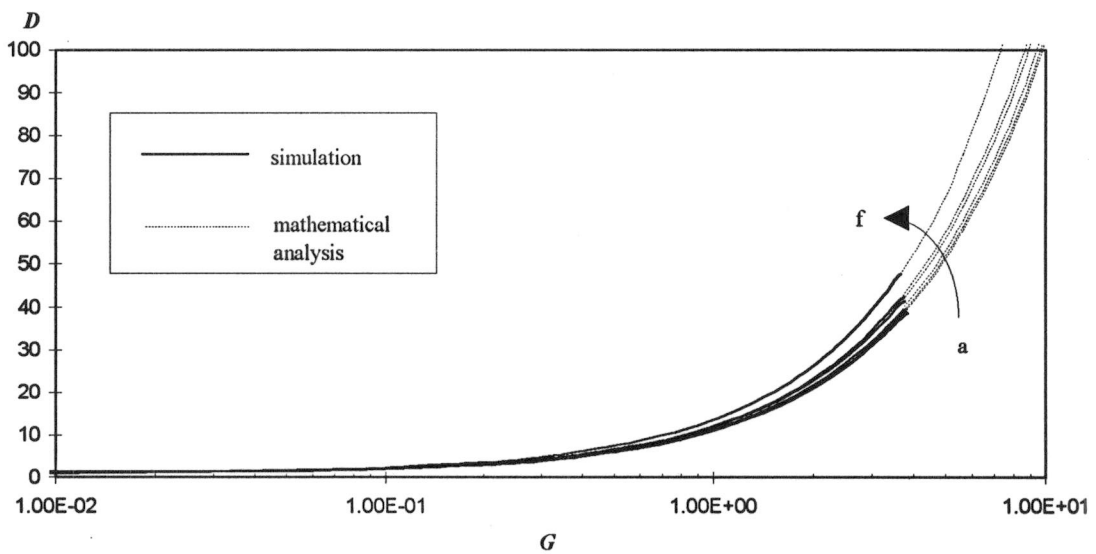


Fig. 6.6b: Delay versus offered traffic using simulation and mathematical analysis of unslotted protocol, varying the inhibit delay fraction  $d$ ; a..f:  $d=0.01, 0.02, 0.04, 0.08, 0.1$  and  $0.2$ ; 8 codes

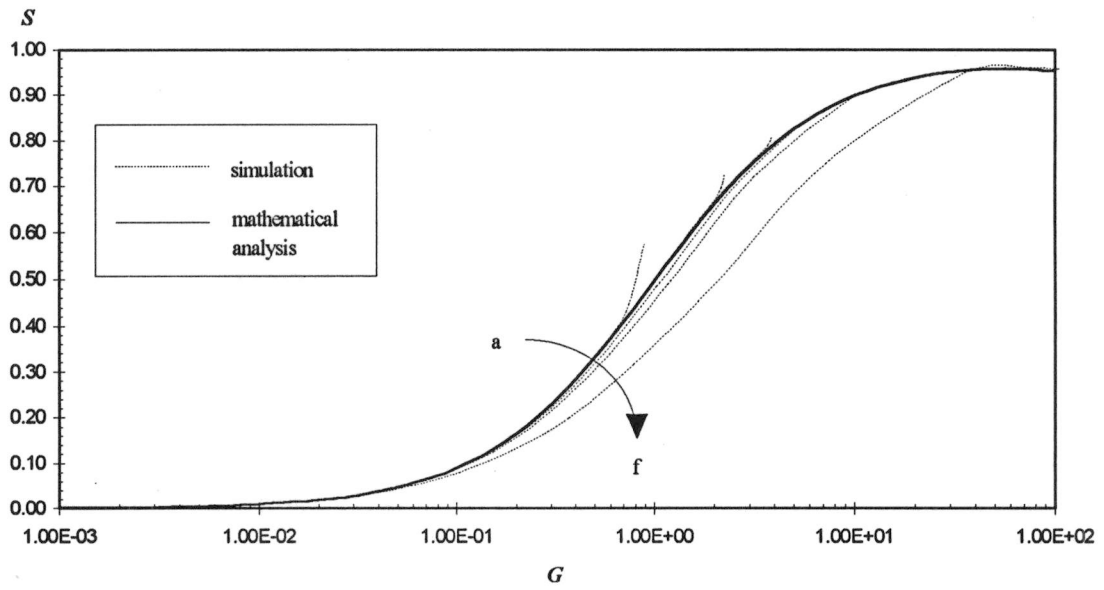


Fig. 6.7a: Throughput versus offered traffic using simulation and mathematical analysis of unslotted protocol, varying  $\alpha$ , the random delay parameter; a..f:  $\alpha=0.01, 0.05, 0.1, 0.5, 1$  and 5; 8 codes

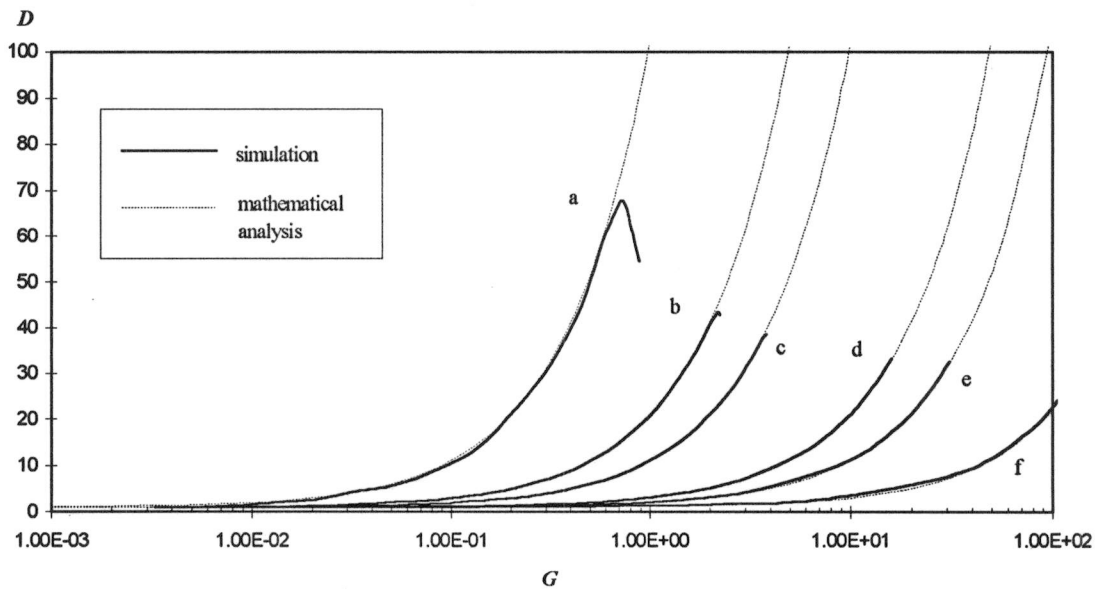


Fig. 6.7b: Delay versus offered traffic using simulation and mathematical analysis of unslotted protocol, varying  $\alpha$ , the random delay parameter; a..f:  $\alpha=0.01, 0.05, 0.1, 0.5, 1$  and 5; 8 codes



Fig. 6.7a and b show the effect of  $\alpha$ , the parameter for the random delay mechanism, on the performance of the unslotted protocol. This parameter is explained in paragraph 4.2.3. If a packet has to be retransmitted, then on average it has to wait  $1/\alpha$  before it can try again. So, for low values of  $\alpha$ , the delay increases tremendously. This can be observed in Fig. 6.7b. In case of the mathematical model, the throughput in Fig. 6.7a does not show much variation if the parameter  $\alpha$  is changed. Fig. 6.7a and b also show strange bendings at the end of simulations. We try to depict this phenomenon more clearly by showing the delay versus throughput in Fig. 6.8.

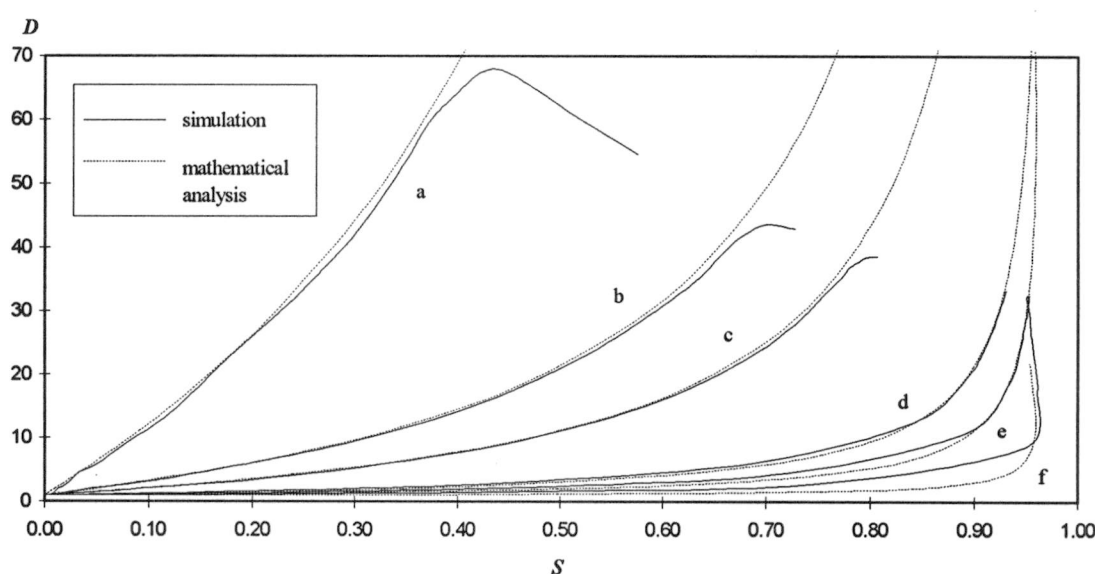


Fig. 6.8: Delay versus throughput using simulation and mathematical analysis of unslotted protocol, varying  $\alpha$ , the random delay parameter; a..f:  $\alpha=0.01, 0.05, 0.1, 0.5, 1$  and  $5; 8$  codes

Fig. 6.8 depicts the delay-throughput characteristics of the unslotted protocol for varying values of  $\alpha$ . If the throughput increases, then the delay also increases. For low values of  $\alpha$  and especially at the end of the curves, the computer simulation does not agree with the mathematical analysis. Especially for curve a, where the delay even decreases tremendously at the end. It is very likely that this is caused by the abrupt program termination and the low value of  $\alpha$ . A low value of  $\alpha$  means that the random delay is large. This means that the collection of statistics has been done while most of the terminals still have to wait a random delay.

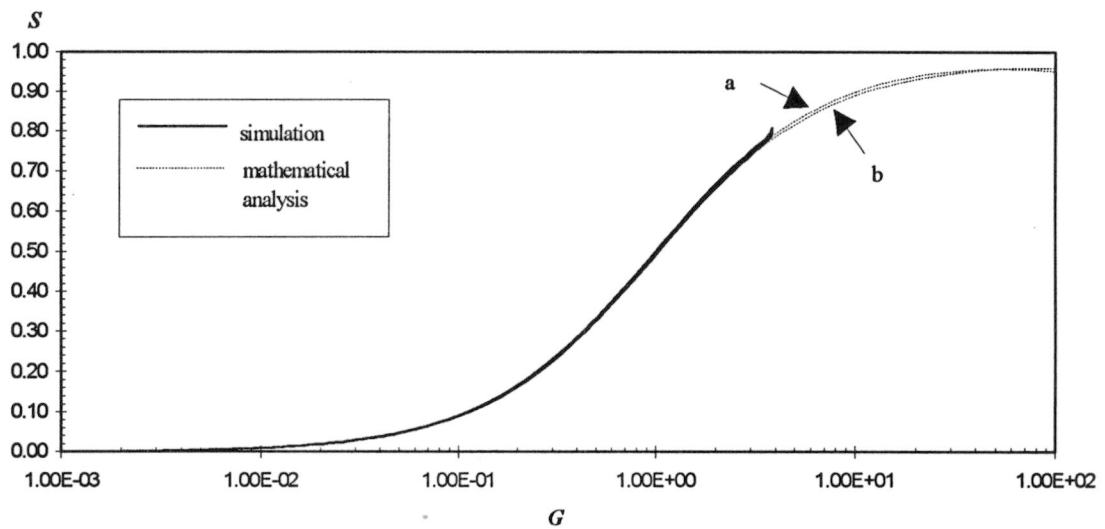


Fig. 6.9a: Throughput versus offered traffic using simulation and mathematical analysis of unslotted protocol, varying the code length; a..b: code length = 31 and 127; 8 codes

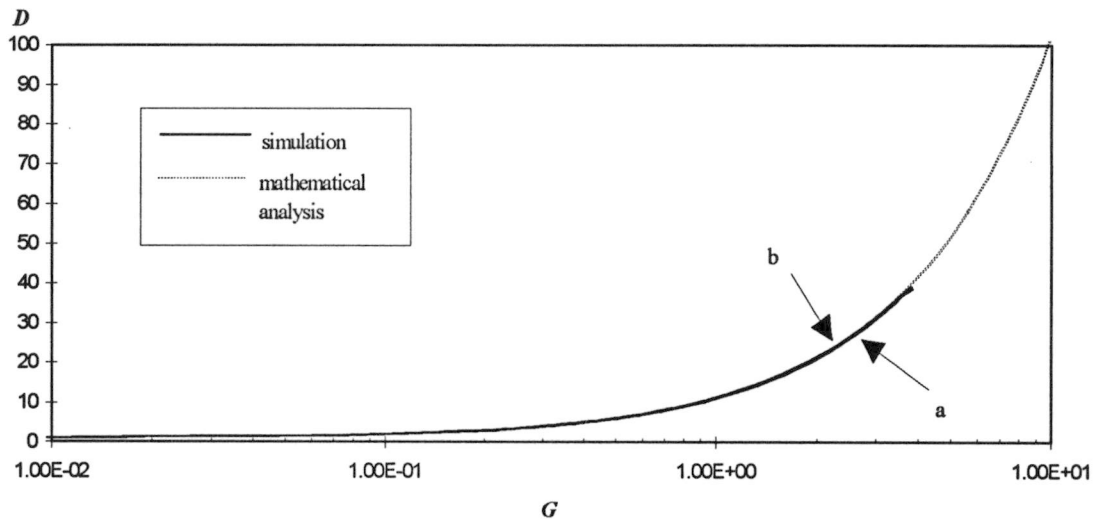


Fig. 6.9b: Delay versus offered traffic using simulation and mathematical analysis of unslotted protocol, varying the code length; a..b: code length = 31 and 127; 8 codes

Fig. 6.9a and b show the effect of code length on the performance of the unslotted protocol. At first sight, it is surprisingly to notice that there is hardly a difference between the performance if the code length is altered. We would expect that the performance of a longer code is better than a shorter one. This is the case for good parameters.

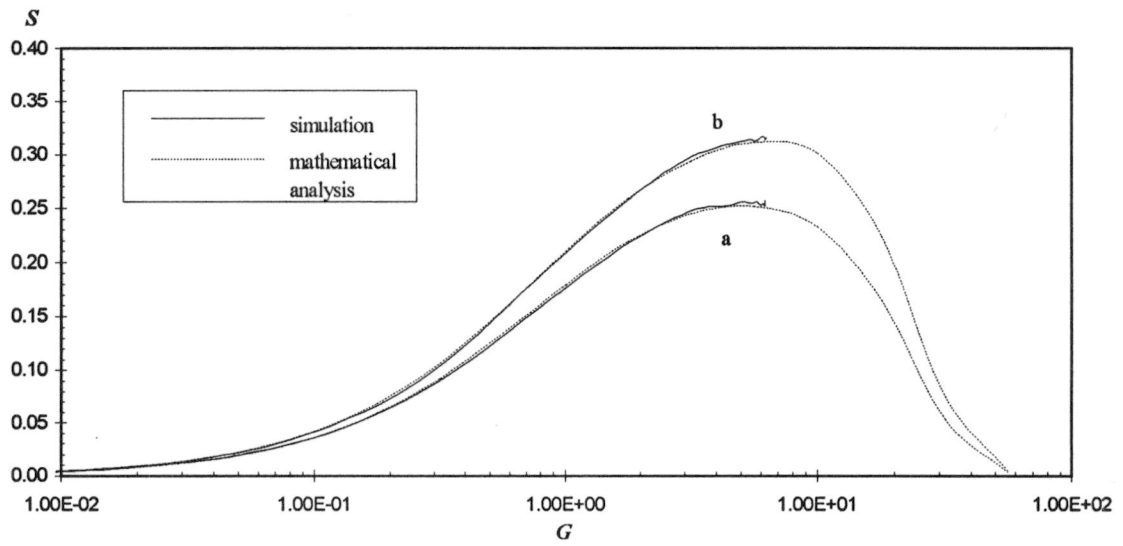


Fig. 6.10a: Throughput versus offered traffic using simulation and mathematical analysis of unslotted protocol, varying the code length; a..b: code length = 31 and 127; 8 codes;  $d=0.2$ ;  $\alpha=0.2$ ; SNR=6 dB

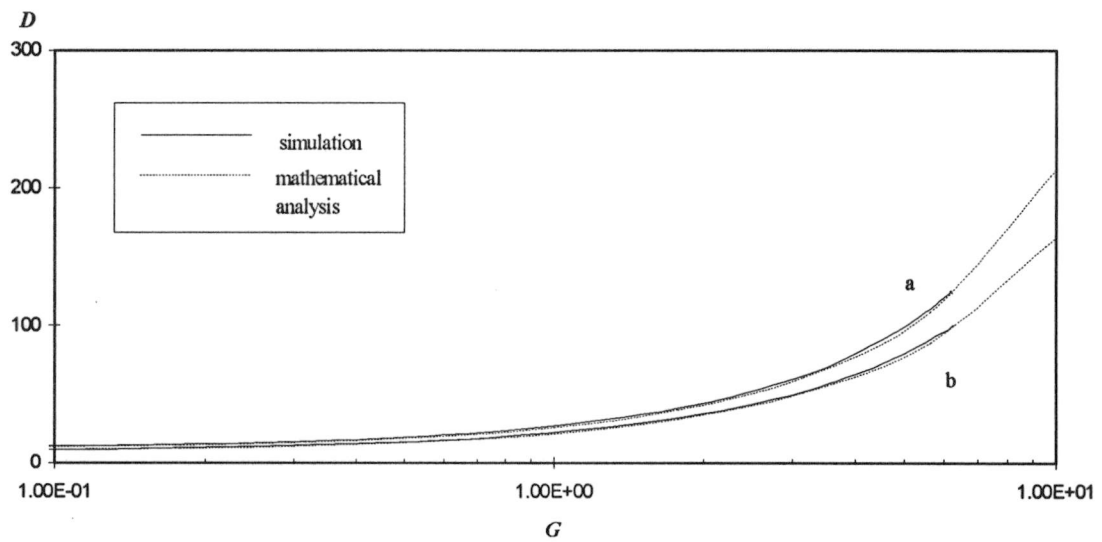


Fig. 6.10b: Delay versus offered traffic using simulation and mathematical analysis of unslotted protocol, varying the code length; a..b: code length = 31 and 127; 8 codes;  $d=0.2$ ;  $\alpha=0.2$ ; SNR=6 dB

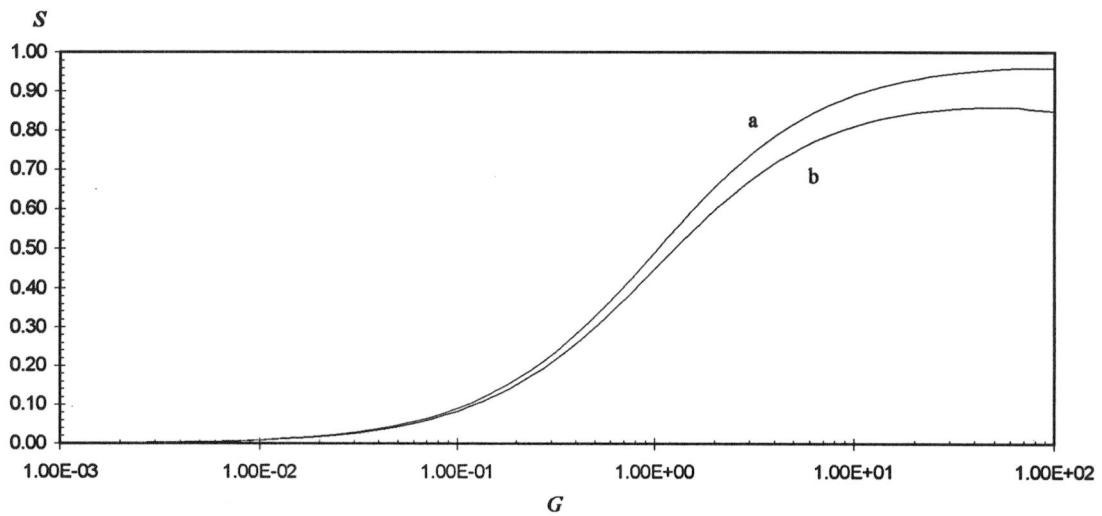


Fig. 6.11a: Throughput versus offered traffic using mathematical analysis of unslotted protocol, varying the code length; a: code length = 127 and bit rate =  $256 \times 1024$ ; b: code length = 31 and bit rate =  $256 \times 1024 \times 4$ ; 8 codes

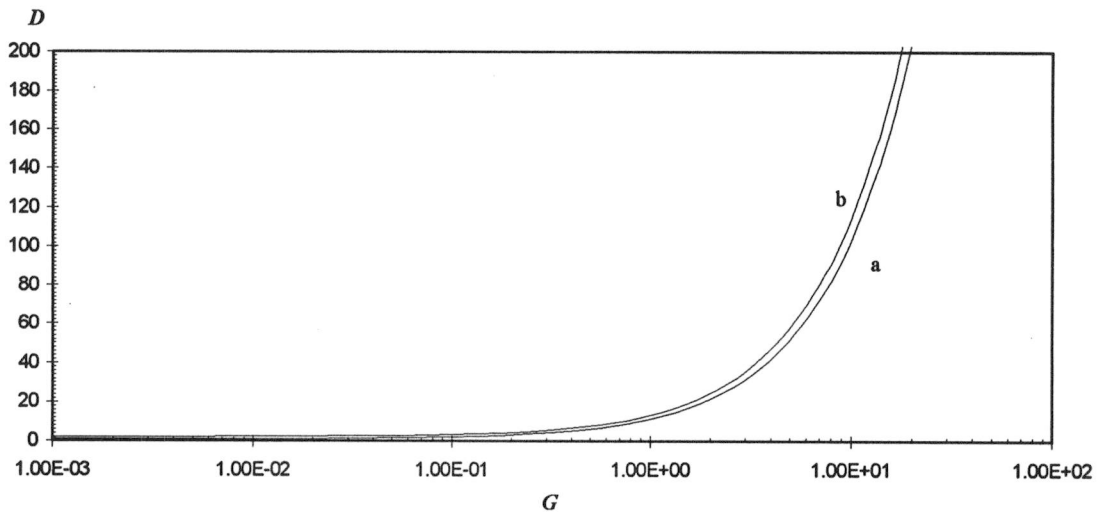


Fig. 6.11b: Delay versus offered traffic using mathematical analysis of unslotted protocol, varying the code length; a: code length = 127 and bit rate =  $256 \times 1024$ ; b: code length = 31 and bit rate =  $256 \times 1024 \times 4$ ; 8 codes

In case the parameters are poor, we do see differences in the performance if the code length is altered (Fig. 6.10a and b). Longer codes perform better than shorter codes because they

have better cross correlations characteristics. However, longer codes need a larger bandwidth. If we keep the bandwidth fixed, and we increase the code length, than we have to decrease the bit rate. This is illustrated in Fig. 6.11a and b. Here, we see that the performanee is better in case the code length is longer.

## 7. CONCLUSIONS AND RECOMMENDATIONS

### 7.1 Conclusions

The unslotted hybrid CDMA/ISMA protocol has been investigated. The protocol performance is measured in terms of throughput  $S$  and delay  $D$  in relationship with the offered traffic  $G$ . Herefore, we have been able to derive a close form formula for the throughput and delay. Also a simulation program has been developed to compare the results. The following conclusions can be drawn from the previous chapters:

- First of all it can be concluded that the results from simulation and mathematical model agree very well with each other. The mathematical model is very suitable for analysing the hybrid protocol even when strong computing power is not available. This mathematical model allows a trade off between accuracy and computing power or computing time.
- Comparison between the slotted and unslotted protocol shows that for low offered traffic the delay for the unslotted protocol is better than the slotted protocol. For high traffic it is the other way round.
- The most important parameter for the slotted p-persistent scheme,  $P_{tr}$ , has a great influence on the protocol's performance. The choice of this parameter depends on the offered traffic the system generates.
- The hybrid protocol supports code sharing extremely well. For a 32 terminal network, only four codes suffice.
- Longer codes do not perform better than shorter codes if the system is under good conditions (the parameters are good). On the other hand, if the parameters are poor, then longer codes can indeed be used to improve the protocol's performance.

- The retransmission parameter  $\alpha$  influences the average packet delay for a great deal.
- If the SNR increases, the performance also improves. However, beyond 10 dB, the performance hardly improves anymore.
- The inhibit delay fraction  $d$  plays an important role in the unslotted non-persistent ISMA scheme. It is concluded that the proposed unslotted hybrid protocol performs very well in case the propagation delay is not a concern ( $d$  is small). If the propagation delay is too high ( $d$  is large) the performance drops quickly.

## 7.2 Recommendations

- The system model should be extended with buffers. How does this feature influence the protocol's performance?
- Suppose the line of sight path is present. With other words, if the channel is modeled as a Rician fading channel, how does the protocol then perform? And does the Rayleigh fading channel describes our indoor channel sufficiently?
- The offered traffic is assumed to be a Poisson process. This is only valid if the number of terminals is large. The investigation of our system is only restricted to 32 terminals because we lack computer power and simulation time. Future research should pay attention to this matter.
- Error corrections can be included to improve the protocol's performance.
- The hybrid protocol composes of CDMA and ISMA. What about other combinations? How about CDMA and TDMA? We also can think of a protocol in which for low offered traffic, the protocol behaves like a random access protocol and for high offered traffic the protocol has a contentionless characteristic.

- Enhance the analysis of the protocol with the feedback channel and the acknowledgements to see how these aspects influence the protocol performance.





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1975.

## APPENDIX A:

“Unslotted Hybrid CDMA/ISMA Protocol for Indoor Wireless Computer Communications”

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# Unslotted Hybrid CDMA/ISMA Protocol for Indoor Wireless Computer Communications

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**ABSTRACT** - A hybrid Code Division Multiple Access / Inhibit Sense Multiple Access (CDMA/ISMA) protocol has been proposed as an effective multiple access scheme for Indoor Wireless Computer Communications. This new protocol combines the advantages of both CDMA and ISMA into one protocol. On the one hand the ISMA protocol introduces a limitation to the number of simultaneous accesses to the transmission channel. On the other hand the CDMA protocol introduces an improvement to the packet survival chance. *Slotted* hybrid CDMA/ISMA protocol has been reported in [1]. It is shown that the performance of the hybrid protocol is indeed better than CDMA only. In addition, code sharing can be applied to reduce hardware cost. This paper presents the performance analysis of the *unslotted* CDMA/ISMA protocol in order to take more advantage of the strength of the hybrid protocol. The performance comparison between the slotted and unslotted hybrid CDMA/ISMA protocol is evaluated in terms of throughput and delay using computer simulation and mathematical analysis.

## I. INTRODUCTION

Indoor wireless office communication is our main research field. The office system we focus on consists of a building in which users work together in groups. The participants generate terminal traffic. Terminals communicate with each other by radio transmission using a random access protocol. Terminals might not detect each other's transmission in radio communications. It can easily happen that two users are hidden from each other by some obstacle, in which case severe performance degradation results. This is called the *hidden terminal problem*. The introduction of a central base station can alleviate this problem by instructing it to send a busy tone to all participating terminals to forbid new transmissions when a transmission is going on. Still, a situation can occur in which two or more terminals simultaneously start their transmission, resulting in a collision. However, a great reduction in the number of simultaneous transmissions is achieved by the introduction of a central base station. This concept is called ISMA [1]-[4].

If we could somehow increase the survival chances of colliding packets, we could improve protocol performance. The near-far effect is one way to achieve this. A packet may 'capture' the receiver if it is much stronger than its competitors. This can happen when terminals use the same transmission power, but at different distances from the receiver. Because the 'near terminals' have better performance compared to the 'far terminals' (due to the better survival chance of the packets), the near-far effect introduces an unfair element among the terminals. Perfect power control, in which the transmitted power of the terminals are adjusted such that their received powers are all equal, can eliminate the near-far effect described above. The performance of CDMA has been reported in a number of publications, e.g. [5]-[10].

The hybrid CDMA/ISMA protocol combines the advantages of both CDMA and ISMA into one protocol. The advantages of CDMA and ISMA are the improvement of the survival chance of packets and the limitation of contention in the channel. Code sharing can also be applied. This is an important aspect because the number of distinct useful transmission codes is limited, especially for short code length.

In [11] the hybrid protocol combines Direct Sequence CDMA with *slotted* p-persistent ISMA. In this paper, we have investigated the performance of the hybrid CDMA/ISMA protocol using the *unslotted* non-persistent ISMA scheme. It is expected that for low traffic, the delay of the unslotted protocol will improve because when a terminal has a data packet to send, it does not have to wait until the start of the next time slot. In addition, even when data packets collide, there is a probability that the data is received correctly due to the use of CDMA. This is the strength of the hybrid CDMA/ISMA protocol. By using the *unslotted non-persistent* ISMA scheme, it is expected that we take more advantage of this strength.

This paper is organized as follows. Section II gives the protocol description of the unslotted hybrid CDMA/ISMA protocol. The performance analysis is then given in Section III. Section IV shows the assumptions used. Results are discussed in Section V. Finally, conclusions can be found in Section VI.

## II. PROTOCOL DESCRIPTION

Fig. 1 depicts an example of the 16-terminal network configuration of the protocol. There is one central base station that is connected to several transceivers by wire. The base station controls the traffic flow with a busy tone that can be detected by all participating terminals.

Because all terminals can detect the busy tone, the hidden terminal problem is solved. Several terminals together with one transceiver form a group. Wireless communication is considered to take place between the transceiver and the terminals. The terminals around each transceiver share the same code. In this way the number of codes can be reduced. When a data packet is ready for transmission, a terminal transmits the packet to its transceiver according to the chosen ISMA scheme.

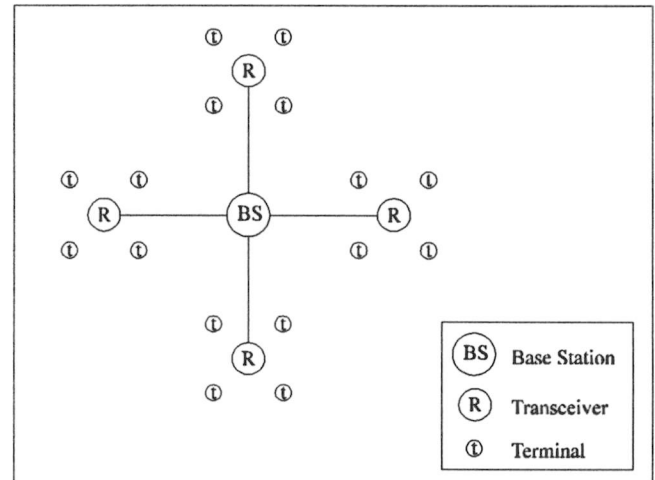


Fig. 1: 16-terminal network configuration

After the arrival of the data packet, the transceiver then simply forwards this packet to the base station which forwards it to all terminals via the transceivers. The destination then receives the packet and decides whether the packet is errorless or not. In case the packet was erroneous, a retransmission must take place. The return channel is not included in our analysis because only the base station makes use of this return channel and therefore contention is not a concern. The state of the terminals can either be free or blocked. At the beginning, the terminals are in the free state. If a packet arrives at a free terminal, the terminal jumps into the blocked state. In the blocked state, the terminal takes care that the arriving data packet is serviced successfully. In the meantime the blocked terminal ignores all incoming packets. This is a consequence of the assumption that the terminals do not have buffers for the incoming data packets.

### *The unslotted non-persistent ISMA protocol*

Users behind terminals generate data packets. The data packets arrive at terminals, which will take care of the correct delivery of the packets. We assume that the arrivals of these packets are generated by a Poisson process. The SDL diagram of the unslotted non-persistent ISMA protocol is shown in Fig. 2. Before transmitting the data packet to the receiver, the terminal first listens to the channel to detect whether there is a busy tone going on. The base station broadcasts a busy



tone to signal that the channel is busy because there is a transmission in progress. If the channel is busy, the terminal waits a random delay before it can try again. Otherwise, the packet will be transmitted immediately. If a collision occurs from the transmission, the collided packets have to wait a random delay before they are allowed to try again (Fig. 2).

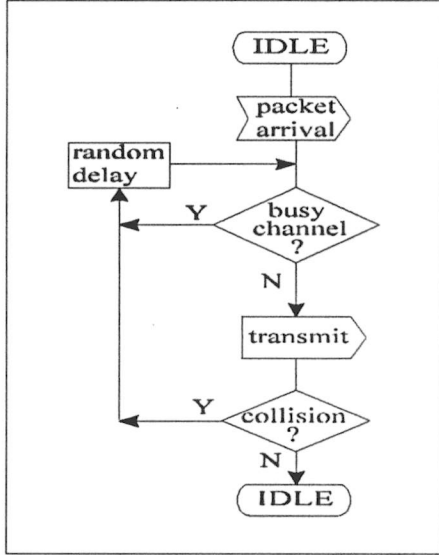


Fig. 2: Unslotted non-persistent ISMA

Although the number of collisions is greatly reduced with ISMA, collisions still can occur. In this unslotted ISMA scheme, collisions may occur due to multiple terminals transmitting packets during an interval called the *inhibit delay fraction*  $d$ . This interval  $d$  is necessary to switch from 'idle' to 'busy'. The reverse interval from 'busy' to 'idle' is denoted by  $d'$ . The inhibit delay fraction is normalized to the packet length, resulting in  $0 \leq d < 1$  (Fig. 3). In this paper we assume  $d = d'$ .

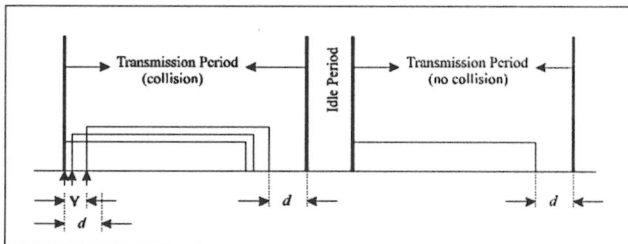


Fig. 3: Data Cycle

### III. PERFORMANCE ANALYSIS

The protocol performance has been measured in terms of throughput  $S$  and delay  $D$  in relationship with the offered traffic  $G$ . The throughput measures the efficiency of the protocol and is defined as the fraction of time in which correct data packets are received. The delay is the time

between the arrival of the first bit of a data packet at a terminal and its arrival of the last bit at the destination.

#### A. Throughput

The performance analysis of the hybrid unslotted non-persistent CDMA/ISMA will be explained from Fig. 3. The time axis in Fig. 3 is normalized to the packet duration. If a packet arrives and the terminal senses the channel free then the packet is sent immediately. Suppose that this time instant is  $t$ . It takes  $d$ , the inhibit delay fraction, before a busy tone is heard by all terminals. This moment occurs at the same time at all terminals because of the symmetric network configuration. Any other packet arriving between  $t$  and  $t+d$  will sense the channel free (because the busy tone has not arrived yet) and will be transmitted resulting in a conflict. Because a CDMA scheme is also used, the conflict does not necessarily result in the loss of all packets. There is a probability that the first arriving packet can still be recovered successfully. If no other terminal transmits a packet during this period  $d$ , then no conflict occurs. We model the channel between the terminals and the corresponding transceivers as a multipath Rayleigh-fading channel. So, when no conflicts occur, there is still a probability that the packet *cannot* be recovered successfully.

Let  $t+Y$  be the time of occurrence of the last packet arriving between  $t$  and  $t+d$ . This obviously means that  $Y$  must be between zero (the first transmitted packet is the only packet in the transmission period) and  $d$  (the end of the vulnerable period). The transmission of all packets arriving in  $(t, t+Y)$  will be completed at  $t+Y+1$ . As noted before, the channel is sensed unused only a period  $d$  later. So, any terminal becoming ready between  $t+d$  and  $t+Y+1+d$  will sense the channel busy and hence will reschedule its packet. The interval between  $t$  and  $t+Y+1+d$  is called a *transmission period (TP)*. There will be *at most* one successful transmission during a TP. The *idle period (IP)* is defined as the period of time between two consecutive TP's. A transmission period plus the following idle period constitute a *cycle*. Let  $\overline{TP}$  be the expected duration of the transmission period,  $\overline{I}$  the expected duration of the idle period, and the average cycle time can be written as:

$$\overline{t_c} = \overline{TP} + \overline{I} \quad (1)$$

Let  $U$  denote the time during a cycle in which the inbound channel (towards base station) is used to carry a successful packet transmission and  $\overline{U}$  the corresponding average value, then we can write the throughput as [12]:

$$S = \frac{\bar{U}}{t_c} \quad (2)$$

The expected useful time  $\bar{U}$  can easily be computed. When a packet is successful, the channel carries useful information for a duration of  $T_{pd}$ , the packet duration. In the unsuccessful case no useful information is carried at all or, in other words:

$$U = \begin{cases} T_{pd} & \text{Successful period} \\ 0 & \text{Unsuccessful period} \end{cases} \quad (3)$$

If  $P_{success}$  denotes the probability that a transmitted packet is successful then

$$\bar{U} = T_{pd} \cdot P_{success} \quad (4)$$

As noted before, the time is normalized to  $T_{pd}$  (the packet duration), and therefore  $T_{pd}$  equals to one. So, this gives us:

$$\bar{U} = P_{success} \quad (5)$$

To calculate the average idle period we make the assumption that the total rate at which users schedule new and retransmitted packets forms a Poisson process with parameter  $G$ . So, new and rescheduled packets arrive at a rate of  $G$  packets per unit time. In literature,  $G$  is also called the offered channel traffic.

The probability of the idle time being greater than some value  $t$  is the probability that no packets are scheduled within a time interval of duration  $t$  and with the assumed Poisson packet scheduling process this probability becomes  $e^{-Gt}$ . Therefore the average value of  $I$  can be expressed as:

$$\bar{I} = \frac{1}{G} \quad (6)$$

The average duration of a transmission period equals to:

$$\bar{TP} = 1 + \bar{Y} + d \quad (7)$$

where  $\bar{Y}$  is the expected value of  $Y$ . Since  $Y$  denotes the time at which the last interfering packet is scheduled, the probability of  $Y$  being smaller than some time  $y$  is the probability that no other packets (either new or retransmissions) are scheduled for transmission in an interval of duration  $d-y$ . With Poisson arrivals, the distribution for  $Y$  is  $e^{-G(d-y)}$ . The average of  $Y$  is therefore given by:

$$\bar{Y} = d - \frac{1}{G} (1 - e^{-Gd}) \quad (8)$$

Applying the formulas obtained above we get:

$$\begin{aligned} \bar{t}_c &= \bar{TP} + \bar{I} = 1 + \bar{Y} + d + \bar{I} \\ &= 1 + \left[ d - \frac{1}{G} (1 - e^{-Gd}) \right] + d + \frac{1}{G} \end{aligned} \quad (9)$$

$$\begin{aligned} S &= \frac{\bar{U}}{\bar{t}_c} = \frac{P_{success}}{1 + \left[ d - \frac{1}{G} (1 - e^{-Gd}) \right] + d + \frac{1}{G}} \\ &= \frac{G \cdot P_{success}}{G(1 + 2d) + e^{-Gd}} \end{aligned} \quad (10)$$

The term  $P_{success}$  in (10) is the only term that needs to be specified. Herefore, we distinguish between four types of transmissions during a transmission period:

1. Successful transmission without conflict
2. Unsuccessful transmission without conflict
3. Successful transmission with conflict
4. Unsuccessful transmission with conflict

If the channel is used for the delivering of a successful packet during a  $TP$  then this is denoted by a successful transmission (situation 1 or 3). The difference between a transmission with or without conflict can be found in the number of packet transmissions during a  $TP$ . In case there is more than one packet transmission during a  $TP$ , we speak of a transmission with conflict. If there is only one packet transmission then the transmission is conflict-free. Keeping this in mind, we can divide  $P_{success}$  into two parts:

$$P_{success} = P_{success|noconflict} + P_{success|conflict} \quad (11)$$

In which  $P_{success|noconflict}$  corresponds to situation 1 and  $P_{success|conflict}$  to situation 3.

$$P_{success|noconflict} = e^{-Gd} P_{ps} \quad (12)$$

This is equal to the probability that no terminal transmits during the inhibit delay fraction  $d$  multiplied by the packet success probability  $P_{ps}$ . The multiplication with  $P_{ps}$  is necessary because of the assumption of the Rayleigh fading channel.

$$P_{ps} = \left( 1 - P_{be|x=0} \right)^{L_p} \quad (13)$$

$P_{be|x=0}$  is the bit error probability in Rayleigh fading channel using the CDMA scheme and is caused by self interference due to multipath.  $L_p$  denotes the packet length. We assume that during the packet duration, the bit error rate remains constant. The calculation of  $P_{success|noconflict}$  is not very complicated. However, this is not the case for  $P_{success|conflict}$  which is the probability that the first packet has arrived successfully given a collision (between two or more packets).

$$P_{success|conflict} = \sum_{x=1}^{K-1} P_x(d) \cdot P\{first\_packet\_OK|x\} \quad (14)$$

in which:

$K$  is the number of active users in the system;  
 $P_x(d)$  is the probability of  $x$  arrivals during  $d$  for a Poisson distribution;  
 $x$  denotes the number of arrivals during  $d$ .

Consider a reference terminal  $T_R$ . This terminal sends its data packet at an idle period first. Other terminals sending within the inhibit delay fraction interfere with the reception at this reference terminal  $T_R$ . The bit error rate depends on the configuration. So, if we try to take the different configurations into account, we have to calculate the bit error probability for all possible configurations and then average.

$$P\{first\_packet\_OK|x\} = E[1 - P_{be|\bar{x}}]^{L_p} \quad (15)$$

$E[.]$  denotes the expectation value, and  $\bar{x}$  a certain configuration of  $x$ . The total number of combinations for  $\bar{x}$  equals  $\binom{K-1}{x}$ . Because we assume an uniform distribution for all possible configurations we can write ( $K$  is the total number of active users in the system):

$$E[1 - P_{be|\bar{x}}]^{L_p} = \frac{1}{\binom{K-1}{x}} \cdot \sum_{\bar{x}} (1 - P_{be|\bar{x}})^{L_p} \quad (16)$$

$$P_{success|conflict} = \sum_{x=1}^{K-1} P_x(d) \cdot \frac{1}{\binom{K-1}{x}} \cdot \sum_{\bar{x}} (1 - P_{be|\bar{x}})^{L_p} \quad (17)$$

The formula for the bit error probability with DPSK modulation in Rayleigh fading channel is given by Kavehrad and Ramamurthi (formula 13a in [13]) in

which the signal to noise ratio (SNR) is taken to be 20 dB. Combining the above results, the throughput can easily be calculated.

### B. Delay

The average packet delay is defined as the duration between the transmission of the first bit till the correct reception of the last bit. For the calculation of the packet delay of the unslotted hybrid CDMA/ISMA protocol, we use the block diagram depicted in Fig. 4.

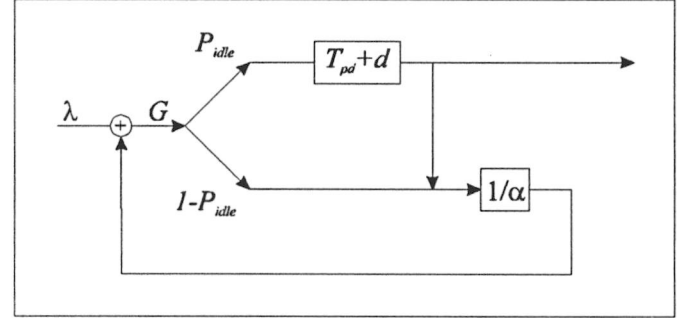


Fig. 4: Delay in the unslotted non-persistent hybrid CDMA/ISMA protocol

If a new packet arrives, the terminal immediately senses the channel to decide if it can transmit the packet. In case the channel is sensed busy (probability  $1 - P_{idle}$ ), the packet is rescheduled for a later time instant. This random delay process is assumed to be negative exponentially distributed with parameter  $\alpha$ . The average random delay time is then  $1/\alpha$ . The average number of times the packet has to suffer this delay is  $G(1 - P_{idle})/S$ . In which  $G$ ,  $S$  and  $P_{idle}$  are defined as before.

On the other hand, if the channel is sensed idle (probability  $P_{idle}$ ), the packet will be transmitted immediately. This transmission will take a packet duration  $T_{pd}$  plus the inhibit delay fraction  $d$  before the knowledge about the outcome of the transmission is available (see also Fig. 3). So, the total delay until the terminal knows if the transmission was successful or not is  $(T_{pd} + d)$ . The transmission of a data packet does not necessarily result in a successful transmission; there are two possibilities: a successful or a failed transmission. If the transmission was successful, the packet leaves the system. The delay this packet suffers is  $(T_{pd} + d)$ . If the transmission was a failure, the packet again has to wait a random delay. In this case, the average delay is  $(T_{pd} + d + \frac{1}{\alpha})$  and the average number of schedulings is  $(\frac{GP_{idle}}{S} - 1)$ . Combining the above results, the average delay is finally given by formula (18).

$$D = \left( \frac{GP_{idle}}{S} - 1 \right) \left[ T_{pd} + d + \frac{1}{\alpha} \right] + \frac{G(1-P_{idle})}{S} \cdot \frac{1}{\alpha} + (T_{pd} + d) \quad (18)$$

#### IV. ASSUMPTIONS

After DPSK modulation the signal is spread with a Gold code. The length of these codes are chosen to be 31. The signal is then transmitted, modulated on a 1.7 GHz carrier. The data rate is arbitrarily chosen to be 256·1024 b/s (0.26 Mb/s). The packet length is 64 bits. The delay spread is supposed to be 100 ns. Perfect power control is assumed to assure that signals from terminals within the same group arrive at the transceiver with the same power.

The far field model [14] is chosen to describe the signal attenuation. Given the transmitted power  $P_T$ , the received power  $P_R$  can be expressed by the following equation:

$$P_R = \frac{g_T g_R}{\left( \frac{4\pi fl}{c} \right)^\alpha} P_T \quad (19)$$

where  $g_T$  and  $g_R$  are respectively the transmitter and the receiver gains,  $f$  the signal frequency,  $l$  the distance between the transmitter and the receiver,  $c$  the speed of light and  $\alpha$  is the attenuation parameter. This attenuation parameter is chosen equal to 2, corresponding to free space propagation.

Fading is the result of the propagation of the transmitted signal through several paths. The channel is modeled as a Rayleigh fading channel. The Rayleigh fading channel model is valid when each path contributes the same amount of energy to the composite received signal. A Line of Sight (LOS) path is therefore assumed to be absent.

An example of a 16-terminal network configuration that we adopted in this paper is shown in Fig. 2. The terminals are clustered around distributed transceivers at a fixed distance (5 m). The distributed transceivers are clustered around the base station at a fixed distance (30 m). Those distances are kept fixed for all simulations. The number of terminals within a transceiver group (also called group size) is chosen as a power of two. For a certain fixed number of participated terminals, we have to halve the group size if we want to double the number of codes (the number of codes is also the number of transceivers). The comparison is fair in this manner, because when we want to investigate the effect of the number of codes on the performance, we

have to keep the number of participated terminals fixed. The number of terminals is 32 unless stated otherwise.

#### V. RESULTS

Fig. 5 a and b depict the simulation comparison between the slotted and unslotted hybrid protocol.

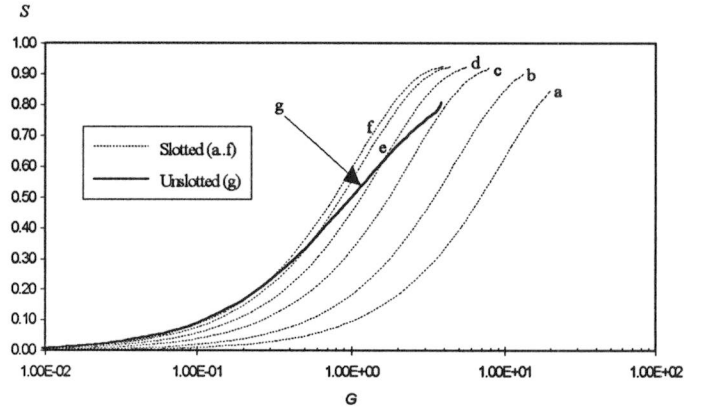


Fig. 5a: Throughput versus offered traffic using simulation of slotted (a.f:  $P_{tr} = 0.1, 0.2, 0.4, 0.6, 0.8, 0.9$ ) and unslotted (g:  $d=0.01$ ) hybrid protocol; 8 codes;  $\alpha=0.1$

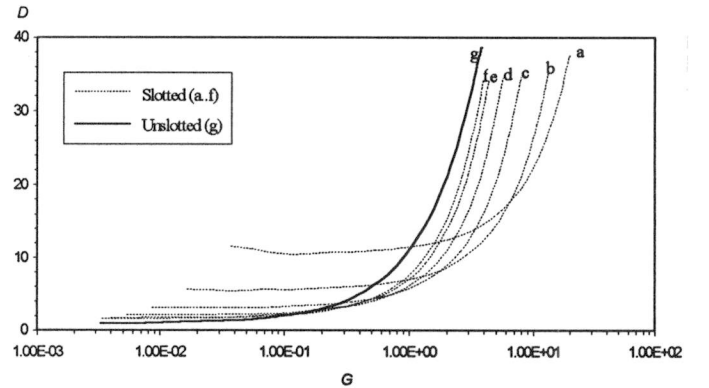


Fig. 5b: Delay versus offered traffic using simulation of slotted (a.f:  $P_{tr} = 0.1, 0.2, 0.4, 0.6, 0.8, 0.9$ ) and unslotted (g:  $d=0.01$ ) hybrid protocol; 8 codes;  $\alpha=0.1$

In [11]  $P_{tr}$  is defined for the slotted protocol as the probability that the packet will actually be transmitted in case the channel is sensed free. For low traffic, the delay of the unslotted protocol is, as expected, better than the slotted protocol. For high traffic the situation is the other way round. The slotted protocol does not differ much from the unslotted protocol for high values of  $P_{tr}$ .

Fig. 6 represents the effect of  $d$  on the performance of the unslotted protocol. For high values of  $d$ , the performance degrades tremendously. This is due to the increase of the number of data packets colliding during

this period. This is why the ISMA protocol is not useful for environment in which the propagation delay is high.

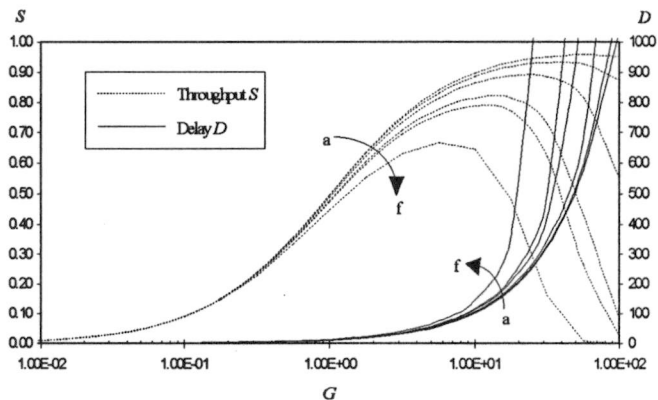


Fig. 6: Throughput and delay versus offered traffic using mathematical analysis of unslotted protocol, varying the inhibit delay fraction  $d$ . a.f:  $d=0.01, 0.02, 0.04, 0.08, 0.1$  and  $0.2$ ; 8 codes;  $\alpha=0.1$

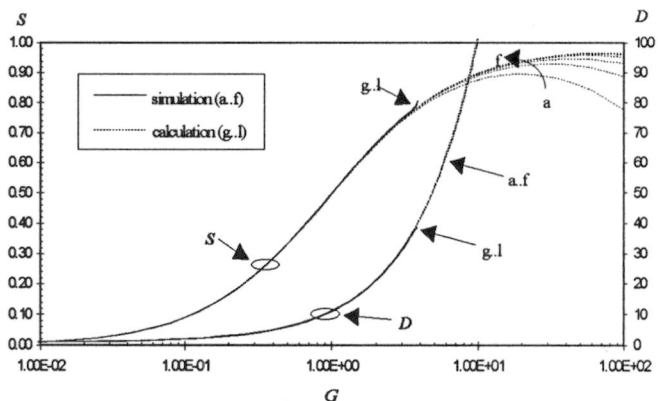


Fig. 7: Comparison between simulation and mathematical analysis for the unslotted protocol. a.f, g.l: 1, 2, 4, 8, 16, 32 codes;  $d=0.01$ ;  $\alpha=0.1$

The comparison between the simulation and mathematical model for the unslotted protocol is given in Fig. 7. In case of simulation, the arrival rate at terminals is the input parameter and  $G$  is one of the output parameters. In addition, due to the assumption that terminals does not have buffers the performance of simulation does not exceed  $G \approx 4$  packets/(packet duration). The simulation and mathematical model does quite agree with each other. This graph also shows the effect of code sharing. For a 32-terminal network, it is sufficient to use only four codes.

## VI. CONCLUSIONS

The unslotted hybrid CDMA/ISMA protocol has been investigated. The protocol performance is measured in terms of throughput  $S$  and delay  $D$  in relationship with

the offered traffic  $G$ . Herefore, we have been able to derive a close form formula for the throughput and delay. Also a simulation program has been developed to compare the results. The results from simulation and mathematical model does quite agree with each other.

Comparison between the slotted and unslotted protocol shows that for low offered traffic the delay for the unslotted protocol is better than the slotted protocol. For high traffic it is the other way round.

Furthermore it is concluded that the proposed hybrid protocol performs very well in case the propagation delay is not a concern. If the propagation delay is too high the performance drops quickly.

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## APPENDIX B:

“Indoor Wireless Computer Communications using Unslotted Hybrid CDMA/ISMA Protocol”

Submitted for participation in the International Conference on Communications (ICC '96).



# Indoor Wireless Computer Communications using Unslotted Hybrid CDMA/ISMA Protocol

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## Presenter

Should the paper be accepted, then Prof. Dr. R. Prasad will attend the conference and also present the paper.

## Technical Subject Areas:

- Computer Communications
- Data Communications
- Broadband Access and Delivery Systems

## Abstract

A hybrid Code Division Multiple Access / Inhibit Sense Multiple Access (CDMA/ISMA) protocol has been proposed as an effective multiple access scheme for Indoor Wireless Computer Communications. This new protocol combines the advantages of both CDMA and ISMA into one protocol. On the one hand the ISMA protocol introduces a limitation to the number of simultaneous accesses to the transmission channel. On the other hand the CDMA protocol introduces an improvement to the packet survival chance. It is shown that the performance of the hybrid protocol is superior to the performance of CDMA only. In addition, code sharing can be applied to reduce hardware cost. To take more advantage of the strength of the hybrid protocol, we have investigated the *unslotted non-persistent* ISMA scheme instead of the *slotted p-persistent* ISMA scheme. The analysis is done for a star-connected multiple access radio network. The performance comparison between the slotted and unslotted hybrid CDMA/ISMA protocol is evaluated in terms of throughput and delay using computer simulation and mathematical analysis.

## I. INTRODUCTION

Indoor wireless communication has proved to be a very attractive means of computer/data communication in an office environment. The office system we focus on consists of a building in which users work together in groups. The participants generate terminal traffic. Terminals communicate with each other by radio transmission using a random access protocol. Terminals might not detect each other's transmission in radio communications. It can easily happen that two users are hidden from each other by some obstacle, in which case severe performance degradation results. This is called the *hidden terminal problem*. The introduction of a central base station can alleviate this problem by instructing it to send a busy tone to all participating terminals to forbid new transmissions when a transmission is going on. Still, a situation can occur in which two or more terminals simultaneously start their transmission, resulting in a collision. However, a great reduction in the number of simultaneous transmissions is achieved by the introduction of a central base station. This concept is called ISMA [1]-[4].

If we could somehow increase the survival chances of colliding packets, we could improve protocol performance. The near-far effect is one way to achieve this. A packet may 'capture' the receiver if it is much stronger than its competitors. This can happen when terminals use the same transmission power, but at different distances from the receiver. Because the 'near terminals' have better performance compared to the 'far terminals' (due to the better survival chance of the packets), the near-far effect introduces an unfair element among the terminals. Perfect power control, in which the transmitted power of the terminals are adjusted such that their received powers are all equal, can eliminate the near-far effect described above. The performance of CDMA has been reported in a number of publications, e.g. [5]-[10].

The hybrid CDMA/ISMA protocol combines the advantages of both CDMA and ISMA into one protocol. The advantages of CDMA and ISMA are the improvement of the survival chance of

packets and the limitation of contention in the channel. Code sharing can also be applied. This is an important aspect because the number of distinct useful transmission codes is limited, especially for short code length.

In [11] the hybrid protocol combines Direct Sequence CDMA with *slotted* p-persistent ISMA. In this paper, we have investigated the performance of the hybrid CDMA/ISMA protocol using the *unslotted* non-persistent ISMA scheme. It is expected that for low traffic, the delay of the unslotted protocol will improve because when a terminal has a data packet to send, it does not have to wait until the start of the next time slot. In addition, even when data packets collide, there is a probability that the data is received correctly due to the use of CDMA. This is the strength of the hybrid CDMA/ISMA protocol. By using the *unslotted non-persistent* ISMA scheme, it is expected that we take more advantage of this strength.

This paper is organized as follows. Section II gives the protocol description of the unslotted hybrid CDMA/ISMA protocol. The performance analysis is then given in Section III. Section IV shows the assumptions used. Results are discussed in Section V. Finally, conclusions are given in Section VI.

## II. PROTOCOL DESCRIPTION

Fig. 1 depicts an example of the 16-terminal network configuration of the protocol. There is one central base station that is connected to several transceivers by wire. The base station controls the traffic flow with a busy tone that can be detected by all participating terminals. Because all terminals can detect the busy tone, the hidden terminal problem is solved. Several terminals together with one transceiver form a group. Wireless communication is considered to take place between the transceiver and the terminals. The terminals around each transceiver share the same code. In this way the number of codes can be

reduced. When a data packet is ready for transmission, a terminal transmits the packet to its transceiver according to the chosen ISMA scheme.

After the arrival of the data packet, the transceiver then simply forwards this packet to the base station which forwards it to all terminals via the transceivers. The destination then receives the packet and decides whether the packet is errorless or not. In case the packet was erroneous, a retransmission must take place. The return channel is not included in our analysis because only the base station makes use of this return channel and therefore contention is not a concern. The state of the terminals can either be free or blocked. At the beginning, the terminals are in the free state. If a packet arrives at a free terminal, the terminal jumps into the blocked state. In the blocked state, the terminal takes care that the arriving data packet is serviced successfully. In the mean time the blocked terminal ignores all incoming packets. This is a consequence of the assumption that the terminals do not have buffers for the incoming data packets.

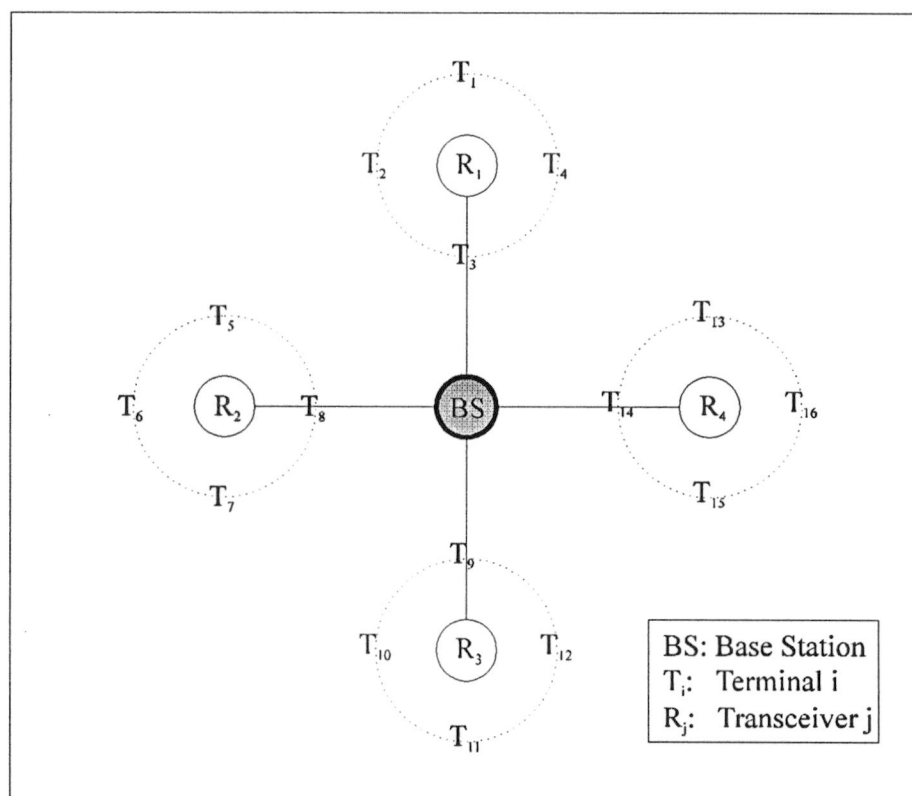


Fig. 1: 16-terminal network configuration

*The unslotted non-persistent ISMA protocol*

Users behind terminals generate data packets. The data packets arrive at terminals, which will take care of the correct delivery of the packets. We assume that the arrivals of these packets are generated by a Poisson process. The Specification and Description Language (SDL) diagram of the unslotted non-persistent ISMA protocol is shown in Fig. 2. Before transmitting the data packet to the receiver, the terminal first listens to the channel to detect whether there is a busy tone going on. The base station broadcasts a busy tone to signal that the channel is busy because there is a transmission in progress. If the channel is busy, the terminal waits a random delay before it can try again. Otherwise, the packet will be transmitted immediately. If a collision occurs from the transmission, the collided packets have to wait a random delay before they are allowed to try again (Fig. 2).

Although the number of collisions is greatly reduced with ISMA, collisions still can occur. In this unslotted ISMA scheme, collisions may occur due to multiple terminals transmitting packets during an interval called the *inhibit delay fraction*  $d$ . This interval  $d$  is necessary to switch from 'idle' to 'busy'. The reverse interval from 'busy' to 'idle' is denoted by  $d'$ . The inhibit delay fraction is normalized to the packet length, resulting in  $0 \leq d < 1$  (Fig. 3). In this paper we assume  $d' = d$ .

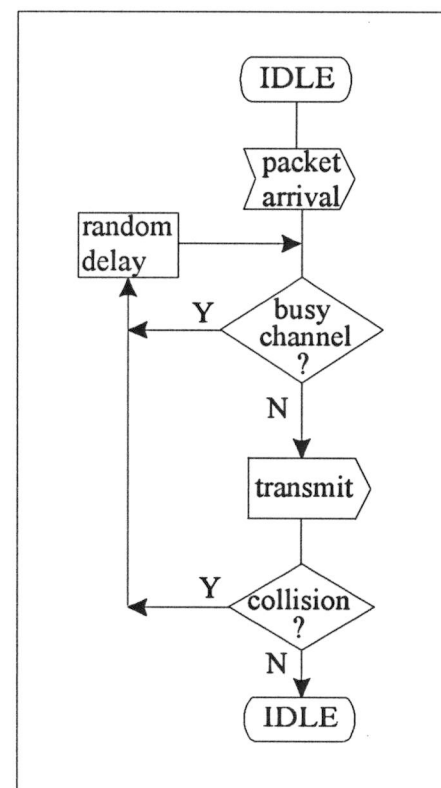


Fig. 2: Unslotted non-persistent  
ISMA

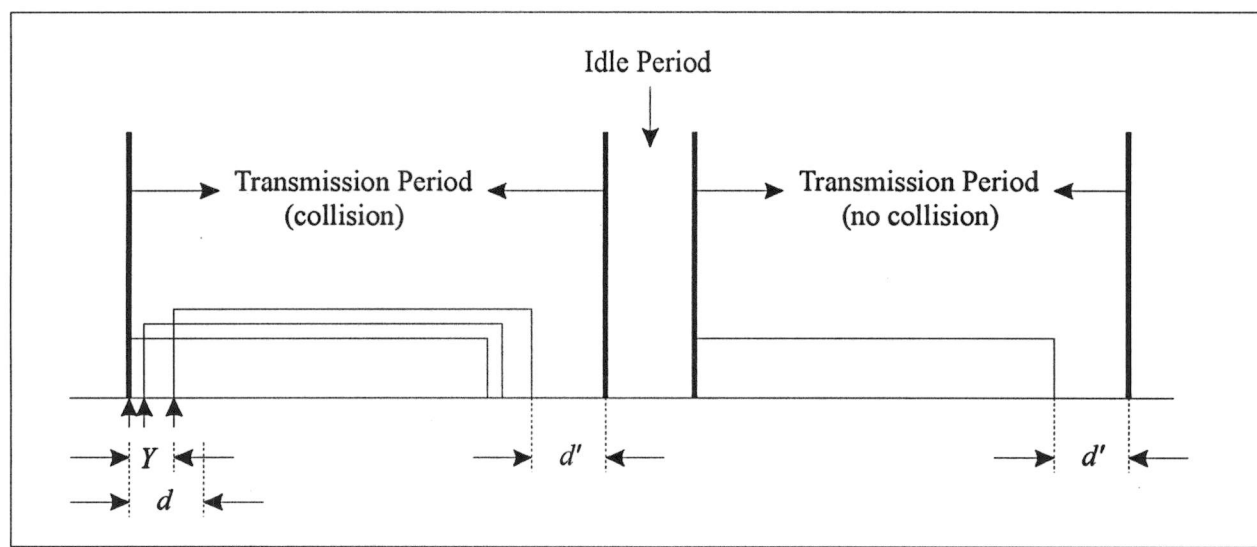


Fig. 3: Data Cycle

### III. PERFORMANCE ANALYSIS

The protocol performance has been measured in terms of throughput  $S$  and delay  $D$  in relationship with the offered traffic  $G$ . The throughput measures the efficiency of the protocol and is defined as the fraction of time in which correct data packets are received. The delay is the time between the arrival of the first bit of a data packet at a terminal and its arrival of the last bit at the destination.

#### A. Throughput

The performance analysis of the hybrid unslotted non-persistent CDMA/ISMA will be explained from Fig. 3. The time axis in Fig. 3 is normalized to the packet duration. If a packet arrives and the terminal senses the channel free, then the packet is sent immediately. Suppose that this time instant is  $t$ . It takes  $d$ , the inhibit delay fraction, before a busy tone is heard by all terminals. This moment occurs at the same time at all terminals because of the symmetric network configuration. Any other packet arriving between  $t$  and  $t+d$  will sense the channel free (because the busy tone has not arrived yet) and will be

transmitted resulting in a conflict. Because a CDMA scheme is also used, the conflict does not necessarily result in the loss of all packets. There is a probability that the first arriving packet can still be recovered successfully. If no other terminal transmits a packet during this period  $d$ , then no conflict occurs. We model the channel between the terminals and the corresponding transceivers as a multipath Rayleigh-fading channel. So, when no conflicts occur, there is still a probability that the packet *cannot* be recovered successfully.

Let  $t+Y$  be the time of occurrence of the last packet arriving between  $t$  and  $t+d$ . This obviously means that  $Y$  must be between zero (the first transmitted packet is the only packet in the transmission period) and  $d$  (the end of the vulnerable period). The transmission of all packets arriving in  $(t, t+Y)$  will be completed at  $t+Y+l$ . As noted before, the channel is sensed unused only a period  $d$  later. So, any terminal becoming ready between  $t+d$  and  $t+Y+l+d$  will sense the channel busy and hence will reschedule its packet. The interval between  $t$  and  $t+Y+l+d$  is called a *transmission period (TP)*. There will be *at most* one successful transmission during a *TP*. The *idle period (IP)* is defined as the period of time between two consecutive *TP*'s. A transmission period plus the following idle period constitute a *cycle*. Let  $\overline{TP}$  be the expected duration of the transmission period,  $\overline{I}$  the expected duration of the idle period, and the average cycle time can be written as:

$$\overline{t_c} = \overline{TP} + \overline{I} \quad (1)$$

Let  $U$  denote the time during a cycle in which the inbound channel (towards base station) is used to carry a successful packet transmission and  $\overline{U}$  the corresponding average value, then we can write the throughput as [12]:

$$S = \frac{\overline{U}}{\overline{t_c}} \quad (2)$$

The expected useful time  $\bar{U}$  can easily be computed. When a packet is successful, the channel carries useful information for a duration of  $T_{pd}$ , the packet duration. In the unsuccessful case no useful information is carried at all or, in other words:

$$U = \begin{cases} T_{pd} & \text{Successful period} \\ 0 & \text{Unsuccessfull period} \end{cases} \quad (3)$$

If  $P_{success}$  denotes the probability that a transmitted packet is successful then

$$\bar{U} = T_{pd} \cdot P_{success} \quad (4)$$

As noted before, the time is normalized to  $T_{pd}$  (the packet duration), and therefore  $T_{pd}$  equals to one. So, this gives us:

$$\bar{U} = P_{success} \quad (5)$$

To calculate the average idle period we make the assumption that the total rate at which users schedule new and retransmitted packets forms a Poisson process with parameter  $G$ . So, new and rescheduled packets arrive at a rate of  $G$  packets per unit time. In literature,  $G$  is also called the offered channel traffic. If the packet arrival times are independent and exponentially distributed with a mean arrival rate of  $G$  packets per second, the probability of  $k$  arrivals in an interval of duration  $t$  is then a Poisson process given by  $P_k(t)$  where

$$P_k(t) = \frac{(Gt)^k \cdot e^{-Gt}}{k!} \quad (6)$$



The probability of the idle time being greater than some value  $t$  is the probability that no packets are scheduled within a time interval of duration  $t$  and with the assumed Poisson packet scheduling process this probability becomes  $e^{-Gt}$ . Therefore the average value of  $I$  can be expressed as:

$$\bar{I} = \frac{1}{G} \quad (7)$$

The average duration of a transmission period equals to:

$$\overline{TP} = 1 + \bar{Y} + d \quad (8)$$

where  $\bar{Y}$  is the expected value of  $Y$ . Since  $Y$  denotes the time at which the last interfering packet is scheduled, the probability of  $Y$  being smaller than some time  $y$  is the probability that no other packets (either new or retransmissions) are scheduled for transmission in an interval of duration  $d-y$ . With Poisson arrivals, the distribution for  $Y$  is  $e^{-G(d-y)}$ . The average of  $Y$  is therefore given by:

$$\bar{Y} = d - \frac{1}{G}(1 - e^{-Gd}) \quad (9)$$

Applying the equations (8) and (9), we get:

$$\bar{t}_c = \overline{TP} + \bar{I} = 1 + \bar{Y} + d + \bar{I} = 1 + \left[ d - \frac{1}{G}(1 - e^{-Gd}) \right] + d + \frac{1}{G} \quad (10)$$

Using (5) and (10), the throughput can be given as:

$$S = \frac{\bar{U}}{t_c} = \frac{P_{success}}{1 + \left[ d - \frac{1}{G}(1 - e^{-Gd}) \right] + d + \frac{1}{G}} = \frac{G \cdot P_{success}}{G(1 + 2d) + e^{-Gd}} \quad (11)$$

The term  $P_{success}$  in (10) is the only term that needs to be specified. Herefore, we distinguish between four types of transmissions during a transmission period:

1. Successful transmission without conflict
2. Unsuccessful transmission without conflict
3. Successful transmission with conflict
4. Unsuccessful transmission with conflict

If the channel is used for the delivering of a successful packet during a  $TP$  then this is denoted by a successful transmission (situation 1 or 3). The difference between a transmission with or without conflict can be found in the number of packet transmissions during a  $TP$ . In case there is more than one packet transmission during a  $TP$ , we speak of a transmission with conflict. If there is only one packet transmission then the transmission is conflict-free. Keeping this in mind, we can divide  $P_{success}$  into two parts:

$$P_{success} = P_{success|noconflict} + P_{success|conflict} \quad (12)$$

In which  $P_{success|noconflict}$  corresponds to situation 1 and  $P_{success|conflict}$  to situation 3.

$$P_{success|noconflict} = e^{-Gd} P_{ps} \quad (13)$$

This is equal to the probability that no terminal transmits during the inhibit delay fraction  $d$  multiplied by the packet success probability  $P_{ps}$ . The multiplication with  $P_{ps}$  is necessary because of the assumption of the Rayleigh fading channel.

$$P_{ps} = \left(1 - P_{be|x=0}\right)^{L_p} \quad (14)$$

$P_{be|x=0}$  is the bit error probability in Rayleigh fading channel using the CDMA scheme and is caused by self interference due to multipath.  $L_p$  denotes the packet length. We assume that during the packet duration, the bit error rate remains constant. The calculation of  $P_{success|noconflict}$  is not very complicated. However, this is not the case for  $P_{success|conflict}$  which is the probability that the first packet has arrived successfully given a collision (between two or more packets).

$$P_{success|conflict} = \sum_{x=1}^{K-1} P_x(d) \cdot P\{first\_packet\_OK|x\} \quad (15)$$

in which:

$K$  is the number of active users in the system;

$x$  denotes the number of arrivals during  $d$ ;

$P_x(d)$  is the probability of  $x$  arrivals during  $d$  for a Poisson distribution and is given by (6).

Consider a reference terminal  $T_R$ . This terminal sends its data packet at an idle period first. Other terminals sending within the inhibit delay fraction interfere with the reception at this reference terminal  $T_R$ . The bit error rate depends on the configuration. So, if we try to take the different configurations into account, we have to calculate the bit error probability for all possible configurations and then average.

$$P\{first\_packet\_OK|x\} = E\left[1 - P_{be|\bar{x}}\right]^{L_p} \quad (16)$$

$E[.]$  denotes the expectation value, and  $\bar{x}$  a certain configuration of  $x$ . The total number of combinations for  $\bar{x}$  equals  $\binom{K-1}{x}$ .

Because we assume an uniform distribution for all possible configurations we can write ( $K$  is the total number of active users in the system):

$$E\left[1 - P_{be|\bar{x}}\right]^{L_p} = \frac{1}{\binom{K-1}{x}} \cdot \sum_{\bar{x}} \left(1 - P_{be|\bar{x}}\right)^{L_p} \quad (17)$$

$$P_{success|conflict} = \sum_{x=1}^{K-1} P_x(d) \cdot \frac{1}{\binom{K-1}{x}} \cdot \sum_{\bar{x}} \left(1 - P_{be|\bar{x}}\right)^{L_p} \quad (18)$$

$$P_{success} = e^{-Gd} \left(1 - P_{be|x=0}\right)^{L_p} + \sum_{x=1}^{K-1} P_x(d) \cdot \frac{1}{\binom{K-1}{x}} \cdot \sum_{\bar{x}} \left(1 - P_{be|\bar{x}}\right)^{L_p} \quad (19)$$

The formula for the bit error probability with DPSK modulation in Rayleigh fading channel is given by Kavehrad and Ramamurthi (formula 13a in [13]) in which the signal to noise ratio (SNR) is taken to be 20 dB. Combining the above results, the throughput can easily be calculated as:

$$S = \frac{G \left( e^{-Gd} \left(1 - P_{be|x=0}\right)^{L_p} + \sum_{x=1}^{K-1} P_x(d) \cdot \frac{1}{\binom{K-1}{x}} \cdot \sum_{\bar{x}} \left(1 - P_{be|\bar{x}}\right)^{L_p} \right)}{G(1+2d) + e^{-Gd}} \quad (20)$$

### B. Delay

The average packet delay is defined as the duration between the transmission of the first bit till the correct reception of the last bit. For the calculation of the packet delay of the unslotted hybrid CDMA/ISMA protocol, we use the block diagram depicted in Fig. 4.

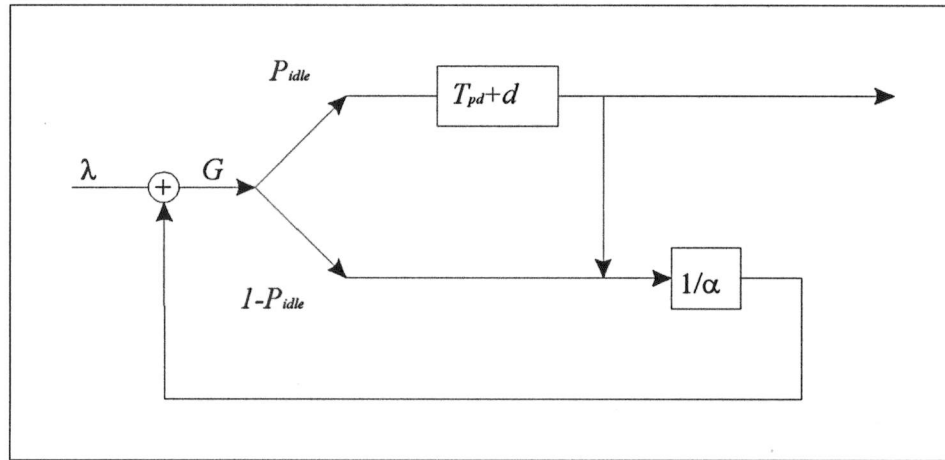


Fig. 4: Delay in the unslotted non-persistent hybrid CDMA/ISMA protocol

If a new packet arrives, the terminal immediately senses the channel to decide if it can transmit the packet. In case the channel is sensed busy (probability  $1 - P_{idle}$ ), the packet is rescheduled for a later time instant. This random delay process is assumed to be negative exponentially distributed with parameter  $\alpha$ . The average random delay time is then  $1/\alpha$ . The average number of times the packet has to suffer this delay is  $G(1 - P_{idle})/S$ .

On the other hand, if the channel is sensed idle (probability  $P_{idle}$ ), the packet will be transmitted immediately. This transmission will take a packet duration  $T_{pd}$  plus the inhibit delay fraction  $d$  before the knowledge about the outcome of the transmission is available (see also Fig. 3). So, the total delay until the terminal knows if the transmission was successful or not is  $(T_{pd} + d)$ . The transmission of a data packet does not necessarily result in a successful transmission; there are two possibilities: a successful or a failed transmission. If the transmission was successful, the packet leaves the system. The

delay this packet suffers is  $(T_{pd} + d)$ . If the transmission was a failure, the packet again has to wait a random delay. In this case, the average delay is  $\left(T_{pd} + d + \frac{1}{\alpha}\right)$  and the average number of schedulings is  $\left(\frac{GP_{idle}}{S} - 1\right)$ . Combining the above results, the average delay is finally given by:

$$D = \left(\frac{GP_{idle}}{S} - 1\right) \left[T_{pd} + d + \frac{1}{\alpha}\right] + \frac{G(1 - P_{idle})}{S} \cdot \frac{1}{\alpha} + (T_{pd} + d) \quad (21)$$

$S$  can be found in formula (20) and  $P_{idle}$  still has to be calculated. The average probability of sensing the channel idle is the average probability that the channel is idle and this probability is equal to the average idle time ( $\bar{I}$ ) divided by the average cycle length ( $\bar{t}_c$ ).

$$P_{idle} = \frac{\bar{I}}{\bar{t}_c} = \frac{\frac{1}{G}}{\frac{1}{G} + \left\{1 + d - \frac{1}{G}(1 - e^{-Gd}) + d\right\}} = \frac{1}{G(1 + 2d) + e^{-Gd}} \quad (22)$$

#### IV. COMPUTER SIMULATION

In this section we describe how the computer simulation program was set up to measure the performance of the unslotted hybrid CDMA/ISMA protocol.

##### A. Representation of Time

In general, if we want to use a computer simulation to analyse a system, the following steps can be distinguished. The dynamic behaviour of the system is studied by tracing various system states as a function of time and then collecting and analysing the system statistics. The events that change the

system state are generated at different points in time, and the passage of time is represented by an internal clock which is incremented and maintained by the simulation program.

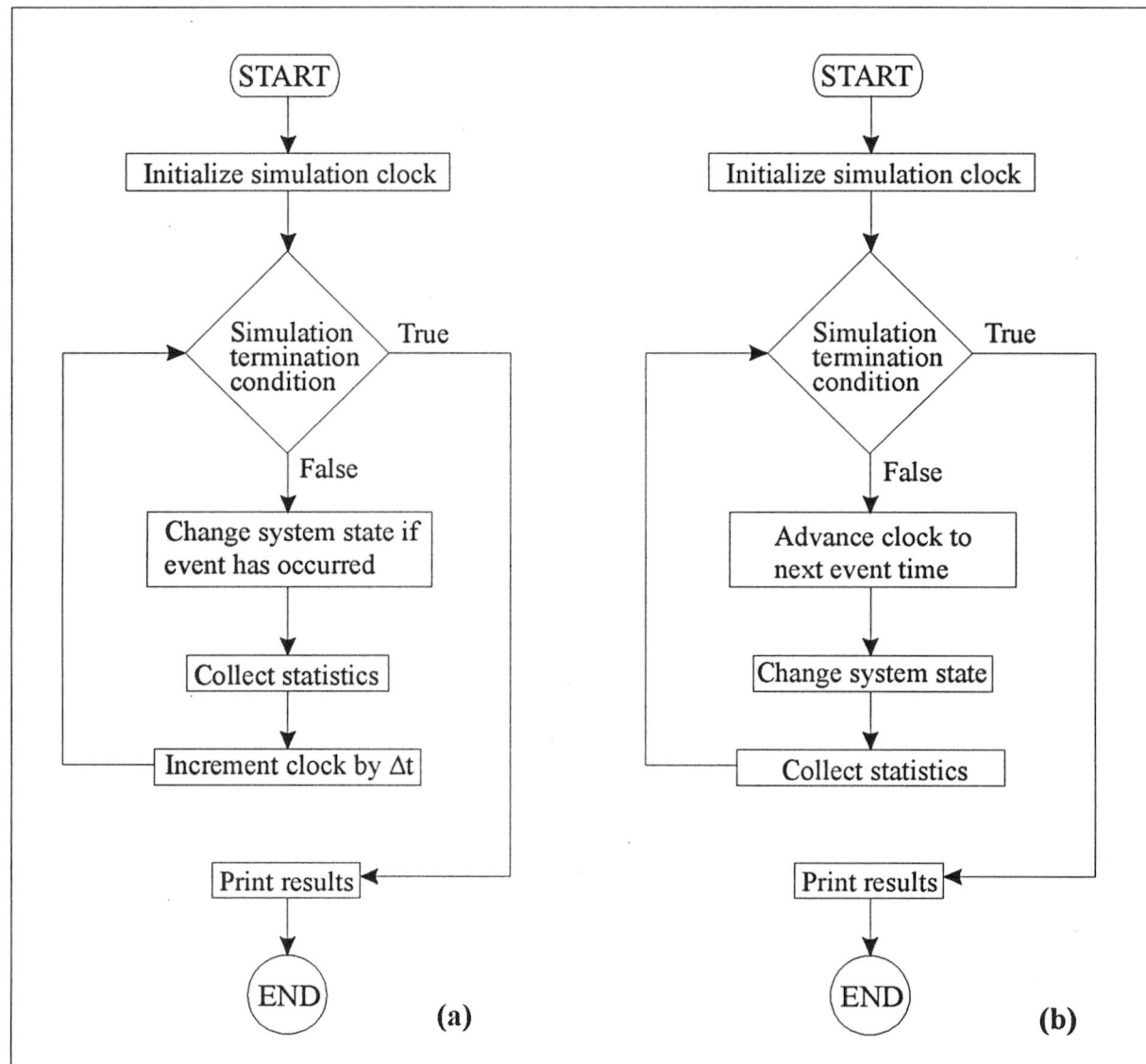


Fig. 5: (a) Interval-oriented simulation; (b) Event-oriented simulation

The simulation time can be advanced in two ways. The first method is the *interval-oriented simulation* (or the uniform time increment method) where the clock is advanced from time  $t$  to  $(t + \Delta t)$  where  $\Delta t$  is a uniform fixed time increment. Fig. 5a depicts this mechanism. The second method is the *event-oriented simulation* (or the variable time increment method) where the clock is incremented from time  $t$

to the next event time  $t'$ , whatever may be the value of  $t'$ . The state changes are made at event time  $t$ , the next event time  $t'$ , and this process is continuously repeated. Thus, only events are represented explicitly in a simulation model and the periods between events are treated as inactive or insignificant and therefore consume no time even though the interevent activities do consume time in the real world (Fig. 5b).

Obviously, method 1 detects the events that occur during the interval  $(t, t + \Delta t)$  only at time  $(t + \Delta t)$ , thereby introducing errors in simulation. Another drawback of this method is that if the interval between two events is very large compared to  $\Delta t$ , then the simulator goes through several unproductive clock increments (during periods of inactivity) and the associated computing effort which will not bring about any change in system states. This fixed time increment method is suitable for the simulation of continuous systems and in particular systems with large numbers of state variables.

The second method involves sorting of event activation times and maintaining *events list*. In our work we will use the event-oriented simulation (Figure 5b) to simulate the unslotted hybrid CDMA/ISMA protocol.

### B. Main Structure

The Program Structure Diagram (PSD) of the main program is depicted in Fig. 6. The simulation program simulates the working of the unslotted hybrid CDMA/ISMA protocol. First of all, the counters have to be reset to the proper values. *Clock* indicates the actual time while the simulation will last *EndTime*. While *Clock* has not achieved *EndTime*, the simulation will loop. The loop is divided in four steps as shown in Fig. 6. In the first step, the program advances *Clock* to the next event time. This event is the first event to occur in future time. Because the event-oriented simulation is used, only events are represented explicitly in the simulation model. Depending on the event type, the program processes this event. This is the second step. The processing can generate other events or change the system state. In the third step, the generated events and the changed system state will be updated. Finally, statistics collection completes the loop.





Fig. 6: PSD of main program

C. Event Lists

From Fig. 6 it emerges that events and event list play an important role in the main program structure.

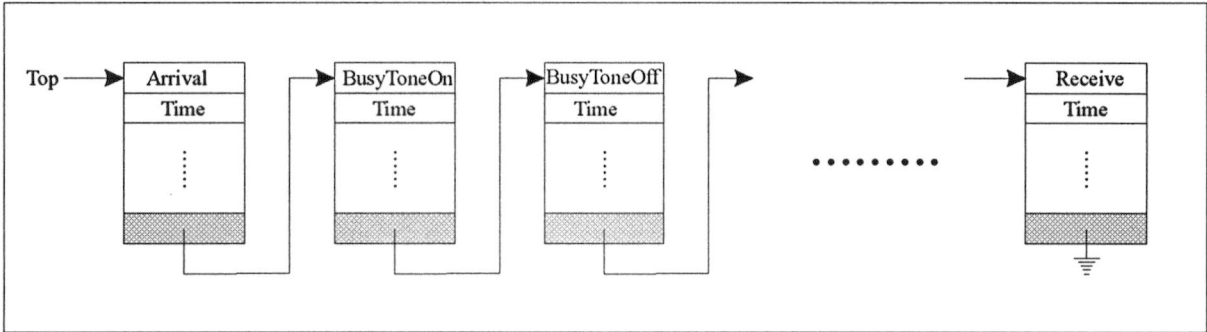


Fig. 7: Example of an event list

The whole program is based on the event list. The event list consists of events linked together in an organised manner. Here, organised means in an order in which the events occur. An example of an event list is shown in Fig. 7. The event list consists of nodes linked together to form a chain. In the event oriented simulation this chain symbolizes the time axis. Every node represents an event and consists of

an *EventType*, the time this event occur, a tail and other information. The tail consists of a pointer pointing to the next node. The last node points to nowhere. This is illustrated with the ground symbol.

#### IV. ASSUMPTIONS

After DPSK modulation the signal is spread with a Gold code. The length of these codes are chosen to be 31. The signal is then transmitted, modulated on a 1.7 GHz carrier. The data rate is arbitrarily chosen to be 256·1024 b/s (0.26 Mb/s). The packet length is 64 bits. The delay spread is supposed to be 100 ns. Perfect power control is assumed to assure that signals from terminals within the same group arrive at the transceiver with the same power.

The far field model [14] is chosen to describe the signal attenuation. Given the transmitted power  $P_T$ , the received power  $P_R$  can be expressed by the following equation:

$$P_R = \frac{g_T g_R}{\left(\frac{4\pi fl}{c}\right)^\alpha} P_T \quad (19)$$

where  $g_T$  and  $g_R$  are respectively the transmitter and the receiver gains,  $f$  the signal frequency,  $l$  the distance between the transmitter and the receiver,  $c$  the speed of light and  $\alpha$  is the attenuation parameter. This attenuation parameter is chosen equal to 2, corresponding to free space propagation.

Fading is the result of the propagation of the transmitted signal through several paths. The channel is modeled as a Rayleigh fading channel. The Rayleigh fading channel model is valid when each path contributes the same amount of energy to the composite received signal. A Line of Sight (LOS) path is therefore assumed to be absent.

An example of a 16-terminal network configuration that we adopted in this paper is shown in Fig. 2. The terminals are clustered around distributed transceivers at a fixed distance (5 m). The distributed transceivers are clustered around the base station at a fixed distance (30 m). Those distances are kept fixed for all simulations. The number of terminals within a transceiver group (also called group size) is chosen as a power of two. For a certain fixed number of participated terminals, we have to halve the group size if we want to double the number of codes (the number of codes is also the number of transceivers). The comparison is fair in this manner, because when we want to investigate the effect of the number of codes on the performance, we have to keep the number of participated terminals fixed. The number of terminals is 32 unless stated otherwise.

## V. RESULTS

Fig. 5 a and b compare the performance of the slotted and unslotted hybrid protocol using computer simulation.

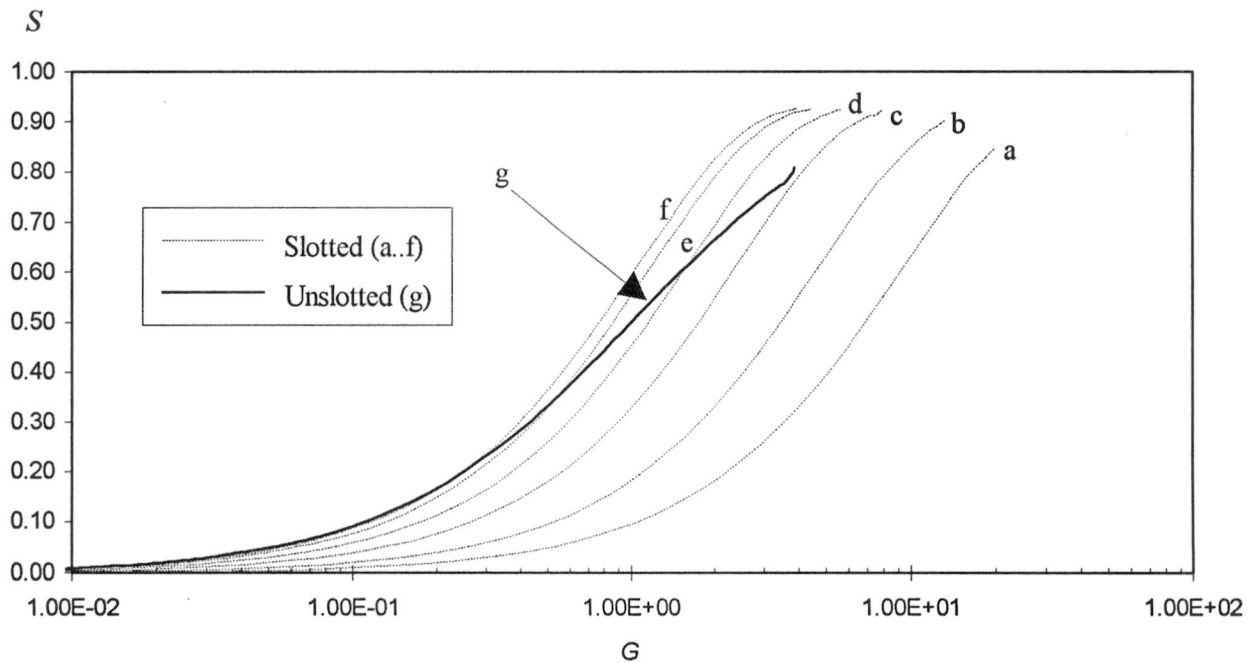


Fig. 5a: Throughput versus offered traffic using simulation of slotted (a..f:  $P_{tr}=0.1, 0.2, 0.4, 0.6, 0.8, 0.9$ ) and unslotted (g:  $d=0.01$ ) hybrid protocol; 8 codes;  $\alpha=0.1$

In [1]  $P_{tr}$  is defined for the slotted protocol as the probability that the packet will actually be transmitted in case the channel is sensed *free*. For low traffic, the delay of the unslotted protocol is, as expected, better than the slotted protocol. For high traffic the situation is the other way round. The slotted protocol does not differ much from the unslotted protocol for high values of  $P_{tr}$ .

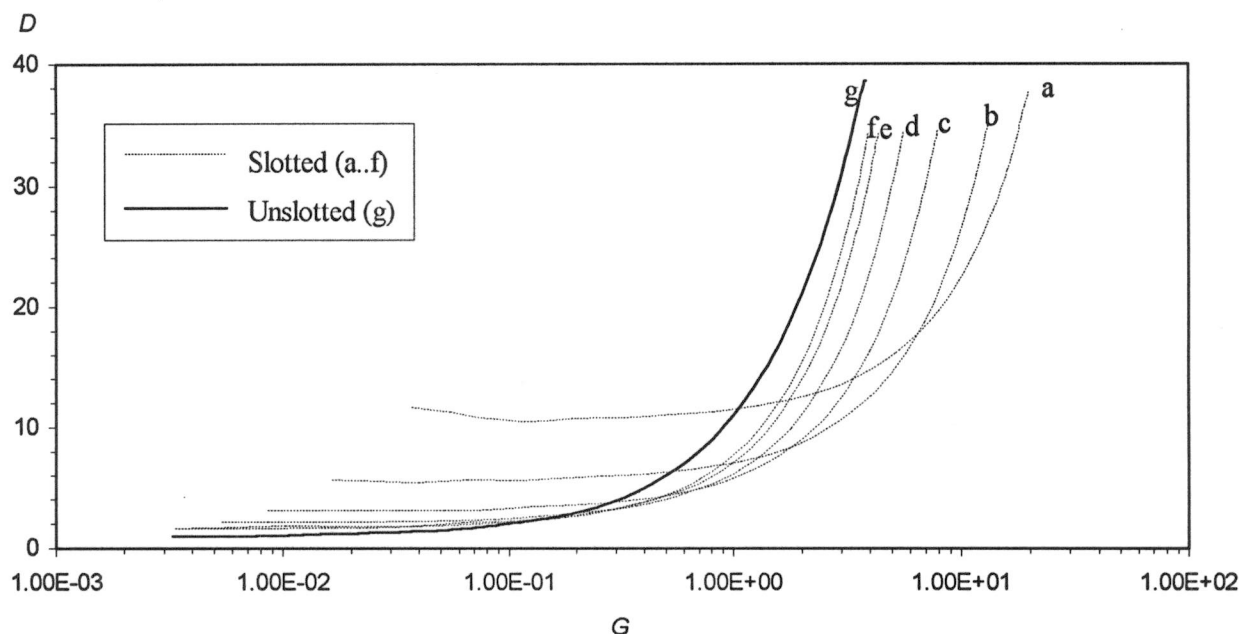


Fig. 5b: Delay versus offered traffic using simulation of slotted (a..f:  $P_{tr}=0.1, 0.2, 0.4, 0.6, 0.8, 0.9$ ) and unslotted (g:  $d=0.01$ ) hybrid protocol; 8 codes;  $\alpha=0.1$

Fig. 6 shows the effect of the inhibit delay fraction  $d$  on the performance of the unslotted protocol. For high values of  $d$ , the performance degrades tremendously. This is due to the increase of the number of data packets colliding during this period. This is why the ISMA protocol is not useful for environment in which the propagation delay is high.

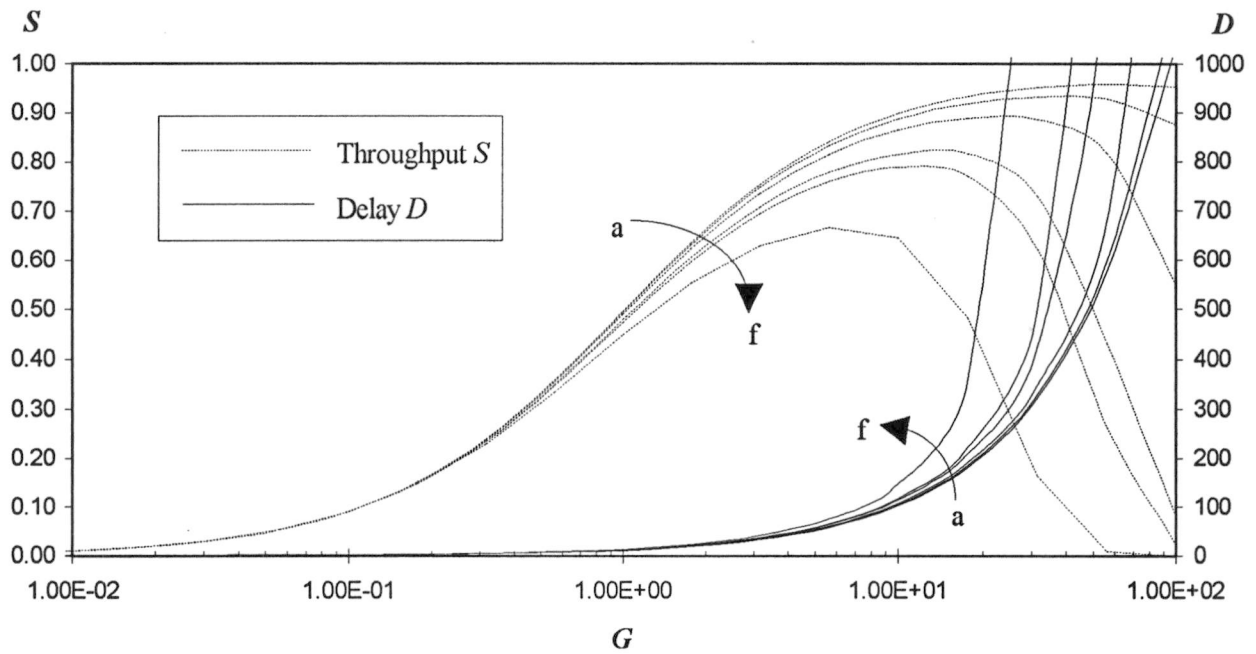


Fig. 6: Throughput and delay versus offered traffic using mathematical analysis of unslotted protocol, varying the inhibit delay fraction  $d$ . a..f:  $d=0.01, 0.02, 0.04, 0.08, 0.1$  and  $0.2$ ; 8 codes;  $\alpha=0.1$

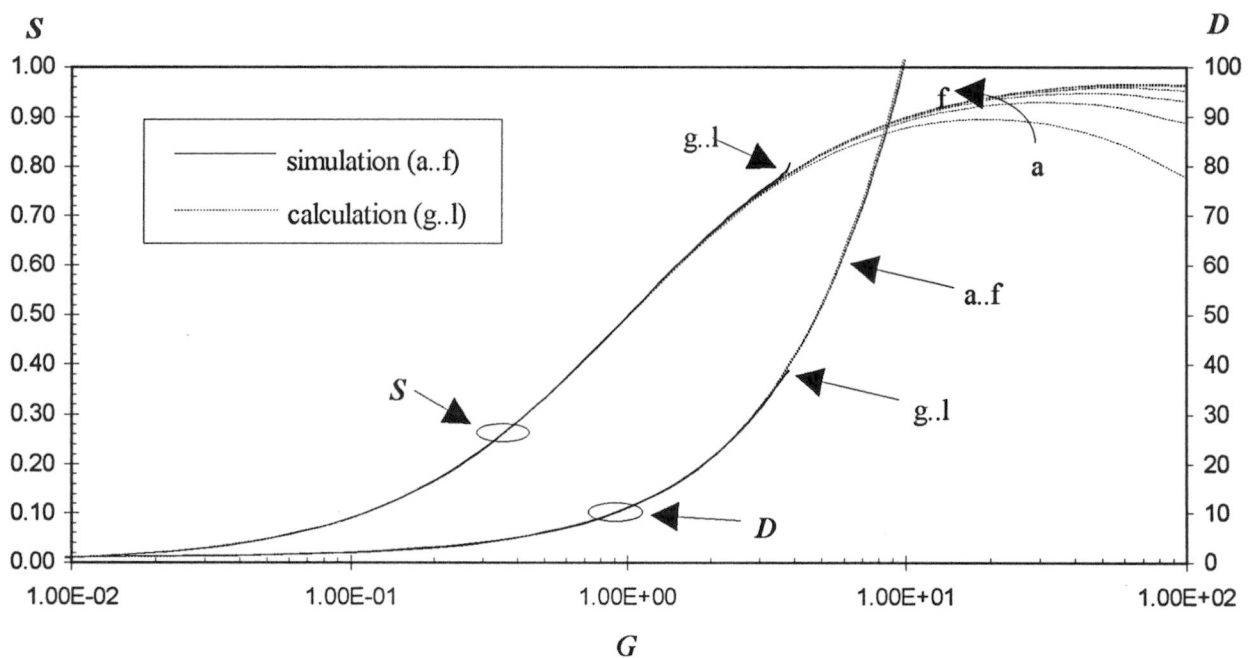


Fig. 7: Comparison between simulation and mathematical analysis for the unslotted protocol. a..f, g..l:

1, 2, 4, 8, 16, 32 codes;  $d=0.01$ ;  $\alpha=0.1$

The comparison between the simulation and mathematical model for the unslotted protocol is given in Fig. 7. In case of simulation, the arrival rate at terminals is the input parameter and  $G$  is one of the output parameters. In addition, due to the assumption that terminals do not have buffers the performance of simulation does not exceed for  $G \approx 4$  packets/(packet duration). The results due to simulation and mathematical model are in good agreement. This graph also shows the effect of code sharing. For a 32-terminal network, it is sufficient to use only four codes.

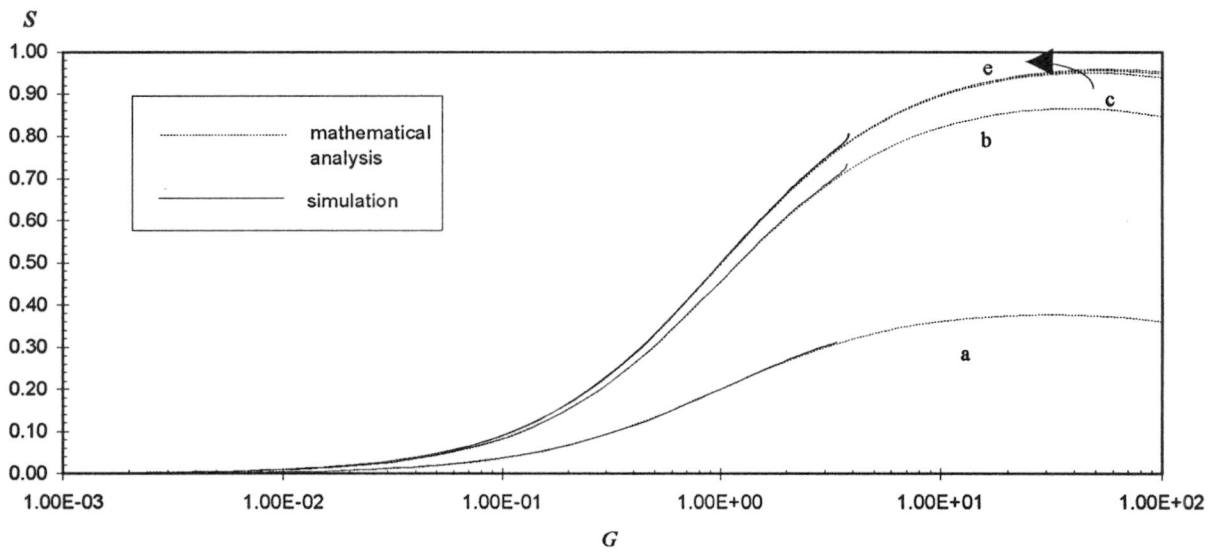


Fig. 8: Comparison between simulation and mathematical analysis for the unslotted protocol and varying SNR; a..e: SNR=6, 8, 10, 12 and 20 dB; number of codes=8;  $d=0.01$ ;  $\alpha=0.1$

Fig. 8 shows the effect of the signal to noise ratio (SNR) on the throughput of the unslotted protocol. Again, simulation and mathematical analysis agree very well with each other. Of course, the throughput improves for increasing SNR. However, Fig. 8 also shows that a SNR of 12 dB suffice. For values of SNR higher than 12 dB, the protocol performance also improves, but only for high values of the offered traffic and the improvement is only additional.

## VI. CONCLUSIONS

The unslotted hybrid CDMA/ISMA protocol has been investigated. The protocol performance is measured in terms of throughput  $S$  and delay  $D$  in relationship with the offered traffic  $G$ . Herefore, we have been able to derive a close form formula for the throughput and delay. Also a simulation program has been developed to compare the results. The results from simulation and mathematical agree very well with each other.

Comparison between the slotted and unslotted protocol shows that for low offered traffic the delay for the unslotted protocol is better than the slotted protocol. For high traffic it is the other way round. Furthermore it is concluded that the proposed hybrid protocol performs very well in case the propagation delay is not a concern. If the propagation delay is too high the performance drops quickly. The results also show that the hybrid protocol supports code sharing extremely good. For a 32 terminal network, only four codes suffice.

Thus it can be concluded that unslotted hybrid CDMA/ISMA protocol will be highly suitable for the indoor wireless computer communication.

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## APPENDIX C:

“Unslotted Hybrid CDMA/ISMA Protocol for Indoor Wireless Computer Communications”

Submitted for publication in IEEE Journal on Selected Areas of Communications.

# Unslotted Hybrid CDMA/ISMA Protocol for Indoor Wireless Computer Communications

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## Abstract

A hybrid Code Division Multiple Access / Inhibit Sense Multiple Access (CDMA/ISMA) protocol has been proposed as an effective protocol for Indoor Wireless Computer Communications. This new protocol combines the advantages of both CDMA and ISMA into one protocol. On the one hand the ISMA protocol introduces a limitation to the number of simultaneous accesses to the transmission channel. On the other hand the CDMA protocol introduces an improvement to the packet survival chance. It is shown that the performance of the hybrid protocol is indeed better than CDMA only. In addition, code sharing can be applied to reduce hardware cost. To take more advantage of the strength of the hybrid protocol, we have investigated the *unslotted non-persistent* ISMA scheme instead of the *slotted p-persistent* ISMA scheme. The performance comparison between the slotted and unslotted hybrid CDMA/ISMA protocol is evaluated in terms of throughput and delay using computer simulation.

## I. INTRODUCTION

On the one hand the Inhibit Sense Multiple Access (ISMA) protocol introduces a limitation to the number of simultaneous accesses to the transmission channel. On the other hand the Code Division Multiple Access (CDMA) protocol introduces an improvement to the packet survival chance. It is therefore to be expected that the combination of the strengths of the two protocols could improve performance. This concept is described below.

Indoor wireless office communication is our main research field. The office system we focus on consists of a building in which users work together in groups. The participants generate terminal traffic. Terminals communicate with each other by radio transmission using a random access protocol. The information is lost when two or more terminals transmit simultaneously, i.e. a collision occurs. Terminals might not detect each other's transmission in radio communications. It can easily happen that two users are hidden from each other by some obstacle, in which case severe performance degradation results. This is called the *hidden terminal problem*. The introduction of a central base station can alleviate this problem by instructing it to send a busy tone to all participating terminals to forbid new transmissions when a transmission is going on. Still, a situation can occur in which two or more terminals simultaneously start their transmission, resulting in a collision. However, a great reduction in the number of simultaneous transmissions is achieved by the introduction of a central base station. This concept is called the Inhibit Sense Multiple Access (ISMA) [1]-[4].

If we could somehow increase the survival chances of colliding packets, we could improve protocol performance. The near-far effect is one way to achieve this. A packet may 'capture' the receiver if it is much stronger than its competitors. This can happen when terminals use the same transmission power, but at different distances from the receiver. Because the 'near terminals' have better performance compared to the 'far terminals' (due to the better survival chance of the packets), the near-far effect introduces an unfair element among the terminals. Perfect power control, in which the transmitted power of the terminals are adjusted such that their received powers are all equal, can eliminate the near-far effect described above. The performance of CDMA has been reported in a number of publications, e.g. [5]-[10].

A hybrid CDMA/ISMA protocol has been proposed in [11]. This new protocol combines the advantages of both CDMA and ISMA into one protocol. The advantages of CDMA and ISMA are the improvement of the survival chance of packets and the limitation of contention in the channel. A less complex receiver structure is needed. More importantly, the number of distinct useful transmission codes is limited, especially for short code length.

In [11] the hybrid protocol combines Direct Sequence CDMA with *slotted* p-persistent ISMA. In this paper, we have investigated the performance of the hybrid CDMA/ISMA protocol using the *unslotted* non-persistent ISMA scheme. It is expected that for low traffic, the delay of the unslotted protocol will improve because when a terminal has a data packet to send, it does not have to wait until the start of the next time slot. In addition, even when data packets collide, there is a probability that the data is received correctly due to the use of CDMA. This is the strength of the hybrid CDMA/ISMA protocol. By using the *unslotted non-persistent* ISMA scheme, it is expected that we take more advantage of this strength.

This paper is organized as follows. Section II gives the protocol description of both the slotted and unslotted hybrid ISMA/CDMA protocol. Assumptions are given in Section III. Computer simulations and results are discussed in Section IV. Finally, conclusions can be found in Section V.

## II. PROTOCOL DESCRIPTION

Figure 1 depicts the network structure of the protocol. There is one central base station that is connected to several receivers by wire. The base station controls the traffic flow with a busy tone that can be detected by all participating terminals. Because all terminals can detect the busy tone, the hidden terminal problem is solved. Several terminals together with one receiver form a group. Wireless communication is considered to take place between the receiver and the terminals. The terminals around each receiver share the same code. In this way the number of codes can be reduced. In this paper we assume that the several groups have the same group size; the distance from the central base station to the several receivers is the same and the distance between the terminals and the receiver within one group is also the same. When a data packet is ready for transmission, a terminal transmits the packet to its receiver according to the chosen ISMA scheme. Two ISMA schemes are important for us and have been discussed later on: *slotted* p-persistent ISMA protocol and *unslotted* non-persistent ISMA protocol.

After the arrival of the data packet, the receiver then simply forwards this packet to the base station which broadcasts it to all terminals. The destination then receives the packet and decides whether the packet is errorless or not. In case the packet was erroneous, a retransmission must take place. The return channel is not included in our analysis because only the base station makes use of this return channel and therefore contention is not a concern. The state of the terminals can either be free or blocked. At the beginning, the terminals are in the free state. If a packet arrives at a free terminal, the terminal jumps into the blocked state. In the blocked state, the terminal takes care that the arriving data packet is serviced successfully. In the mean time the blocked terminal ignores all incoming packets. This

is a consequence of the assumption that the terminals do not have buffers for the incoming data packets. The arrivals of data packets at terminals are assumed to be generated by a Poisson process. This means that the inter-arrival times of data packets at terminals are negative exponentially distributed with  $P_{\text{Arr}}$  as parameter ( $P_{\text{Arr}}$  indicates the average number of arrivals per terminal per time unit).

#### *A. The slotted p-persistent ISMA protocol*

Users behind terminals generate data packets. The data packets arrive at terminals, which will take care of the correct delivery of the packets. We assume that the arrivals of these packets are generated by a Poisson process. The Specification and Description Language (SDL) like diagram of the slotted p-persistent ISMA protocol is depicted in Figure 2. In this case, the terminal uses the slotted p-persistent ISMA protocol to service its data packet when it is in the blocked state. Before transmitting the data packet to the receiver, the terminal first listens to the channel to hear whether there is a busy tone transmitted by the base station. The base station broadcasts a busy tone to signal that the channel is busy because there is a transmission in progress. If there is a busy tone, indicating that the channel is busy, the terminal postpones its transmission to the next time slot. Otherwise, if the channel is free, the terminal performs a random binary experiment to determine whether it will send its data packet. If the terminal decides not to transmit, it waits until the next time slot and draws a number again to see if it is permitted to send. This mechanism is built in to reduce the number of simultaneous transmissions. Simultaneous transmissions cause collisions and must be reduced as much as possible.

If the terminal is permitted to send, it transmits its data packet to the receiver. It is possible that two or more terminals transmit in the same time slot. In this case a collision occurs. Because we also make use of CDMA, a collision does not necessarily result in the loss of the data packets. There is a probability that the receiver can lock onto one of the data streams and this data packet can be received properly. In case the data packet is received erroneously, the terminal then waits for a random period and starts all over again (see also Figure 2). On the other hand, if no collision occurs, the terminal assumes that the packet has been received correctly and leaves the blocked state to jump into the free state. The random delay for retransmissions is set to zero in our experiment because the random binary experiment already introduces a delay after a collision.

#### *B. The unslotted non-persistent ISMA protocol*

The SDL-like diagram of the unslotted non-persistent ISMA protocol is shown in Figure 3. This ISMA protocol is unslotted. There are two major differences between this unslotted and the previously described slotted protocol. The first difference is that the terminal waits a random period when it senses that the channel is busy. Because the time axis is not divided into time slots, terminals have to wait a random period. After the random delay, the terminals are allowed to sense the channel again to see if it



is free. The random period is modeled to be negative exponentially distributed. The second difference is related to the first one. In the unslotted scheme, the terminals immediately transmit if the channel is free. As a consequence of the first difference mentioned above, the terminals already wait a random period in case the channel is busy. So, if the channel is free the number of terminals attempting to transmit is already reduced. It is not necessary to let the terminals draw a random number to see if they may transmit. Although the number of collisions is greatly reduced with ISMA, collisions can still occur. In this unslotted ISMA scheme, collisions may occur due to multiple terminals transmitting packets during an interval called the *inhibit delay fraction* ( $d$ ). This interval  $d$  is necessary to switch from 'idle' to 'busy'. The reverse interval from 'busy' to 'idle' is denoted by  $d'$ . The inhibit delay fraction is normalized to the packet length, resulting in  $0 \leq d < 1$  (Figure 4). In this paper we assume  $d = d'$ .

### III. ASSUMPTIONS

#### A. Transmitter and Receiver

After DPSK modulation the signal is spread with a Gold code. The length of these codes are chosen to be 31 and 127 to evaluate the influence of code length. The signal is then transmitted, modulated on a 1.7 GHz carrier. The data rate is arbitrarily chosen to be 256·1024 b/s (0.26 Mb/s). The packet length is 64 bits. Perfect power control is assumed to assure that signals from terminals within the same group arrive at the receiver with the same power.

We consider one antenna for each receiver. So, only spread spectrum diversity is taken into account. The number of resolvable paths  $L$  can be found with the following formula [12]:

$$L = \left\lfloor \frac{T_m}{T_c} \right\rfloor + 1 \quad (1)$$

where  $T_m$  is the RMS delay spread,  $T_c$  is the chip duration and  $\lfloor x \rfloor$  denotes the largest integer smaller than or equal to  $x$ . The relationship between  $T_c$  the chip duration and  $T_b$  the bit duration is given by the code length  $N$ :

$$T_c = \frac{T_b}{N} \quad (2)$$

We have a number of resolvable paths due to spread spectrum. The maximum value for  $T_m$  has been measured by Saleh and Valenzuela and is 100 ns [13]. For our work we assume  $T_m$  to be 100 ns. For  $N = 31$  this gives one resolvable path, and for  $N = 127$  this gives us four resolvable paths.

### B. Signal Attenuation Model

The far field model is chosen to describe the signal attenuation [14]. Given the transmitted power  $P_T$ , the received power  $P_R$  can be expressed by the following equation:

$$P_R = \frac{g_T g_R}{\left(\frac{4\pi f l}{c}\right)^\alpha} P_T \quad (3)$$

where  $g_T$  and  $g_R$  are respectively the transmitter and the receiver gains,  $f$  the signal frequency,  $l$  the distance between the transmitter and the receiver,  $c$  the speed of light and  $\alpha$  is the attenuation parameter. This attenuation parameter is chosen equal to 2, corresponding to free space propagation.

### C. Rayleigh Fading Channel

Fading is the result of the propagation of the transmitted signal through several paths. The channel is modeled as a Rayleigh fading channel. The Rayleigh fading channel model is valid when each path contributes the same amount of energy to the composite received signal. A Line of Sight (LOS) path is therefore assumed to be absent. The formula for the bit error probability is given by Kavehrad and Ramamurthi [15]:

$$P_{be|\beta_{\max}, \{\tau_{ik}\}, L} = Q(a, b) - \frac{1}{2} \left( 1 + \frac{\mu_{12}}{\sqrt{\mu_1 \mu_2}} \right) I_0(a, b) \exp\left(-\frac{a^2 + b^2}{2}\right) \quad (4.1)$$

where

$Q(a, b)$  is the Marcum Q-function

$$a = \frac{m}{\sqrt{2}} \left| \frac{1}{\sqrt{\mu_1}} - \frac{1}{\sqrt{\mu_2}} \right|, \text{ and } b = \frac{m}{\sqrt{2}} \left| \frac{1}{\sqrt{\mu_1}} + \frac{1}{\sqrt{\mu_2}} \right| \quad (4.2)$$

$$m = A\beta_{\max} T_b b_1^0 = A\beta_{\max} T_b b_1^{-1} \quad (4.3)$$

$$\begin{aligned} \mu_1 = A^2 E \left[ \sum_{k=1}^{Nt} X_k^2 + \hat{X}_k^2 + Y_k^2 + \hat{Y}_k^2 | \{\tau_{lk}\}, L \right] \\ + 2A^2 E [X_1 \hat{X}_1 + Y_1 \hat{Y}_1 | \{\tau_{lk}\}, L] + 2\sigma_n^2 \end{aligned} \quad (4.4)$$

$$\mu_2 = A^2 E \left[ \sum_{k=1}^{Nt} X_k^2 + \hat{X}_k^2 + Y_k^2 + \hat{Y}_k^2 | \{\tau_{lk}\}, L \right] + 2\sigma_n^2 \quad (4.5)$$

$$\mu_{12} = A^2 E \left[ \sum_{k=1}^{Nt} (X_k \hat{X}_k + Y_k \hat{Y}_k) + \hat{X}_1^2 + \hat{Y}_1^2 | \{\tau_{lk}\}, L \right] \quad (4.6)$$

In which  $\sigma_n^2$  is the thermal noise power;  $A$  is the transmitted signal amplitude;  $Nt$  denotes the number of active users;  $L$  is the number of paths;  $\beta_{lk}$  and  $\tau_{lk}$  are the  $l$ th path gain and delay respectively, for the  $k$ th user;  $\beta_{\max}$  denotes the largest value of  $\beta_{ll}$ . The conditional expectations in the above expressions can be evaluated as follows:

$$E[X_1^2 | \{\tau_{lk}\}, L] = E[Y_1^2 | \{\tau_{lk}\}, L] = \sum_{\substack{l=1 \\ l \neq j}}^L R_{11}^2(\tau_{l1}) \rho_{l1} \quad (5.1)$$

$$E[\hat{X}_1^2 | \{\tau_{lk}\}, L] = E[\hat{Y}_1^2 | \{\tau_{lk}\}, L] = \sum_{\substack{l=1 \\ l \neq j}}^L \hat{R}_{11}^2(\tau_{l1}) \rho_{l1} \quad (5.2)$$

$$E[X_1 \hat{X}_1 | \{\tau_{lk}\}, L] = E[Y_1 \hat{Y}_1 | \{\tau_{lk}\}, L] = \sum_{\substack{l=1 \\ l \neq j}}^L R_{11}(\tau_{l1}) \hat{R}_{11}(\tau_{l1}) \rho_{l1} \quad (5.3)$$

$$E[X_k^2 | \{\tau_{lk}\}, L] = E[Y_k^2 | \{\tau_{lk}\}, L] = \sum_{l=1}^L R_{lk}^2(\tau_{lk}) \rho_{lk} \quad (5.4)$$

$$E[\hat{X}_k^2 | \{\tau_{lk}\}, L] = E[\hat{Y}_k^2 | \{\tau_{lk}\}, L] = \sum_{l=1}^L \hat{R}_{lk}^2(\tau_{lk}) \rho_{lk} \quad ; k \geq 2 \quad (5.5)$$

$$E[X_k \hat{X}_k | \{\tau_{lk}\}, L] = E[Y_k \hat{Y}_k | \{\tau_{lk}\}, L] = \sum_{l=1}^L R_{lk}(\tau_{lk}) \hat{R}_{lk}(\tau_{lk}) \rho_{lk} \quad ; k \geq 2 \quad (5.6)$$

In which

$$R_{lk}(\tau) = \int_0^\tau a_k(t - \tau) a_l(t) dt \quad (5.7)$$

$$\hat{R}_{lk}(\tau) = \int_\tau^T a_k(t - \tau) a_l(t) dt \quad (5.8)$$

$$a_k(t) = \sum_i a_k^i P_{T_c}(t - iT_c) \quad (5.9)$$

$$b_k(t) = \sum_{j=-\infty}^{\infty} b_k^j P_T(t - jT_b) \quad (5.10)$$

$$\text{and} \quad \rho_{lk} = \frac{1}{2} E[\beta_{lk}^2] \quad (5.11)$$

#### D. Network Configuration

An example of a 16-terminal network configuration that we adopted in this paper is shown in Figure 5. The terminals are clustered around distributed receivers at a fixed distance (5 m). The distributed receivers are clustered around the base station at a fixed distance (30 m). Those distances are kept fixed for all simulations. The number of terminals within a receiver group (also called group size) is chosen as a power of two. For a certain fixed number of participated terminals, we have to halve the group size if we want to double the number of codes (the number of codes is also the number of receivers). The comparison is fair in this manner, because when we want to investigate the effect of the number of codes on the performance, we have to keep the number of participated terminals fixed.

## IV. COMPUTER SIMULATION AND RESULTS

In this section the performance of the slotted p-persistent and the unslotted non-persistent ISMA scheme is given. The performance analysis for the slotted scheme has been evaluated by computer simulation and Markov model in [11]. In this paper we concentrate on the performance comparison between the slotted and unslotted hybrid CDMA/ISMA protocol through computer simulation.

### A1. Representation of Time

As a first approach to analyse the *unslotted* hybrid CDMA/ISMA protocol, a computer simulation program will be set up. The dynamic behaviour of the system is studied by tracing various system states as a function of time and then collecting and analysing the system statistics. The events that change the system state are generated at different points in time, and the passage of time is represented by an internal clock which is incremented and maintained by the simulation program.

The simulation time can be advanced in two ways. The first method is the *interval-oriented simulation* (or the uniform time increment method) where the clock is advanced from time  $t$  to  $(t + \Delta t)$  where  $\Delta t$  is a uniform fixed time increment. Figure 6a depicts this mechanism. The second method is the *event-oriented simulation* (or the variable time increment method) where the clock is incremented from time  $t$  to the next event time  $t'$ , whatever may be the value of  $t'$ . The state changes are made at event time  $t$ , the next event time  $t'$ , and this process is continuously repeated. Thus, only events are represented explicitly in a simulation model and the periods between events are treated as inactive or insignificant and therefore consume no time even though the interevent activities do consume time in the real world (Figure 6b).

Obviously, method 1 detects the events that occur during the interval  $(t, t + \Delta t)$  only at time  $(t + \Delta t)$ , thereby introducing errors in simulation. Another drawback of this method is that if the interval between two events is very large compared to  $\Delta t$ , then the simulator goes through several unproductive clock increments (during periods of inactivity) and the associated computing effort which will not bring about any change in system states. This fixed time increment method is suitable for the simulation of continuous systems and in particular systems with large numbers of state variables.

The second method involves sorting of event activation times and maintaining *events list*. In our work we will use the event-oriented simulation (Figure 6b) to simulate the hybrid unslotted CDMA/ISMA protocol.

### *A2. Main Structure*

The Program Structure Diagram (PSD) of the main program is depicted in Figure 7. The simulation program simulates the working of the hybrid unslotted CDMA/ISMA protocol. First of all, the counters have to be reset to the proper values. *Clock* indicates the actual time while the simulation will last *EndTime*. While *Clock* has not achieved *EndTime*, the simulation will loop. The loop is divided in four steps as shown in Figure 7. In the first step, the program advances *Clock* to the next event time. This event is the first event to occur in future time. Because the event-oriented simulation is used, only events are represented explicitly in the simulation model. Depending on the event type, the program processes this event. This is the second step. The processing can generate other events or change the system state. In the third step, the generated events and the changed system state will be updated. Finally, statistics collection completes the loop.

### *A3. Event Lists*

From Figure 7 it emerges that events and event lists play an important role in the main program structure. The whole program is based on the event lists. The event lists consist of events linked together in an organised manner. Here, organised means in an order in which the events occur. An example of the event list is shown in Figure 8. The event lists consist of nodes linked together to form a chain. In the event oriented simulation this chain symbolizes the time axis. Every node represents an event and consists of an *EventType*, the time this event occur, a tail and other information. The tail consists of a pointer pointing to the next node. The last node points to nowhere. This is illustrated with the ground symbol.

### *B. Performance Measurements*

The protocol performance has been measured in terms of throughput  $S$  and delay  $D$  in relationship with the arrival rate at terminals  $P_{Arr}$ . Note that in [11] the physical traffic is used instead of  $P_{Arr}$  (see also Figures 2 and 3 for the physical traffic reference). The physical traffic indicates the number of terminals that actually transmit data packets to the receiver.

The throughput  $S$  measures the efficiency of the protocol and is defined as the fraction of time in which correct data packets are received. The delay  $D$  is the time between the arrival of the first bit of a data packet at a terminal and its arrival of the last bit at the destination.

First, the performance inherent to the unslotted non-persistent scheme will be presented. After that, we present the performance comparison between the slotted and unslotted scheme.

### *C. Performance of the unslotted scheme*

As discussed before, the inhibit delay fraction ( $d$ ) is one of the parameters that influences the protocol behavior. By varying this inhibit delay fraction and measuring the performance, Figure 9 and Figure 10 are obtained. Figure 9 depicts the throughput of the unslotted scheme as a function of the arrival rate at terminals for different values of the inhibit delay fraction. This inhibit delay fraction  $d$  is varied between 0.01 to 0.2 (note that  $d$  is normalized and can vary between zero and one). For low arrival rates, the throughput curves hardly differ. This can be explained by the fact that for low arrival rates, the probability that more than one packet arrives within the inhibit delay fraction is very low. On the other hand, for high traffic, the throughput drops for high value of  $d$ . The throughput degradation is due to the collisions of data packets. Because for high traffic, more packets arrive within the inhibit delay fraction if  $d$  is too large.

The same explanation applies to the delay curves of the unslotted scheme which is shown in Figure 10. In this figure, the delay is depicted versus the arrival rate at terminals for different values of the inhibit delay fraction. The delay is normalized in the number of packet durations for comparison purposes with the slotted protocol later on. As can be expected from Figure 9, the delays for high traffic get worse for high values of the inhibit delay fraction  $d$ .

Another parameter which influences the performance of the unslotted hybrid protocol is the random delay that can be found in Figure 3. Before a retransmission (caused by a collision or a busy channel) can take place, terminals have to undergo a random delay first. This is to prevent too many terminals from retransmitting their data packets at the same time. We model this random process to be negative exponentially distributed with  $\lambda$  as parameter.  $\lambda$  represents the mean number of arrivals per packet duration (the time is normalized in packet duration for comparison purposes) and influences the period the data packet has to wait before it can be retransmitted. The average time a packet has to wait before it can be retransmitted is then given by:  $E[\text{random delay}] = 1/\lambda$  [16]. As a result of varying  $\lambda$ , we obtain the performance shown in Figures 11 and 12.

For different values of  $\lambda$ , Figure 11 depicts the throughput of the unslotted scheme as a function of the arrival rate at terminals.  $\lambda = 0.1$  means that on average 0.1 packets per packet duration will become available for retransmission at each blocked terminal (the time is normalized in packet durations). This results in an average delay of 10 packet durations before a retransmission takes place. For low values of  $\lambda$ , the throughput is also low. This is because erroneous packets have to wait a long period before they may be retransmitted. The same explanation also applies to Figure 12 in which the delay is depicted versus the arrival rate.

### *C. Comparison between the slotted and unslotted scheme*

The comparison between the slotted and unslotted hybrid CDMA/ISMA protocol is presented in Figures 13 and 14. The throughput comparison between the slotted and unslotted hybrid protocol is shown in Figure 13. The number of users is kept constant to 32. The parameter we vary in those curves is the number of codes used (this is also the number receivers or groups in the network). The range in which we vary this parameter is from 1 to 32. For low number of codes used, the performance is low for both slotted and unslotted scheme. Low number of codes used means a high number of users shares the same code because the number of users is kept constant. Here we see the effect of code sharing. For low arrival rates, code sharing has not much influence because the probability that the terminals in the same group start sending at the same time is very low.

For high arrival rates, the throughput performance of the unslotted scheme is better than the slotted one. The explanation can be found in the characteristics of the retransmission mechanism for both schemes. For the slotted scheme this is a random binary experiment and for the unslotted scheme this is the Poisson process with the negative exponential distribution. In addition, for the unslotted scheme, data packets can be sent immediately if terminals are allowed to and do not have to wait to the start of a slots as in the slotted scheme. The delay characteristics of both the slotted and unslotted schemes are depicted in Figure 14.

## V. CONCLUSIONS

The protocol performance is measured in terms of throughput  $S$  and delay  $D$  in relationship with the arrival rate at terminals  $P_{Arr}$ . The performance of the unslotted non-persistent hybrid CDMA/ISMA protocol has been investigated by computer simulation. Two decisive parameters which influences the protocol performance have been analyzed: the inhibit delay fraction and the average time a packet has to wait to be retransmitted. The increase of these parameters have a negative impact on both throughput and delay for high arrival rates.

The performance comparison between the slotted p-persistent and the unslotted non-persistent hybrid CDMA/ISMA protocol is also considered. For high traffic, the performance of the unslotted protocol is significantly better than that of the slotted protocol. This is caused by differences in the retransmission algorithms of both protocols. The unslotted protocol also shows superiority compared to the slotted protocol in terms of packets delay. This is because for the unslotted scheme, data packets can be sent immediately if terminals are allowed to and do not have to wait for the start of a slot as in the slotted scheme.



Computer simulations show that the unslotted scheme is the scheme of choice in the hybrid CDMA/ISMA protocol. The unslotted hybrid protocol will be investigated using a Markov model in future work.

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### List of Figure Captions

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- Figure 2: Slotted p-persistent ISMA scheme
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- Figure 7: PSD of main program
- Figure 8: Example of an event list
- Figure 9: Throughput  $S$  versus arrival rate  $P_{Arr}$  for different values of the inhibit delay fraction  $d$  parameter for the unslotted hybrid CDMA/ISMA protocol; Code length = 31; SNR = 20 dB; Number of groups = 8; Group size = 4.
- Figure 10: Delay  $D$  versus arrival rate  $P_{Arr}$  for different values of the inhibit delay fraction  $d$  parameter for the unslotted hybrid CDMA/ISMA protocol; Code length = 31; SNR = 20 dB; Number of groups = 8; Group size = 4.
- Figure 11: Throughput  $S$  versus arrival rate  $P_{Arr}$  for different values of  $\lambda$  for the unslotted hybrid CDMA/ISMA protocol; Code length = 31; SNR = 20 dB; a..f : 0.1, 0.5, 1, 2, 5, 10 for  $\lambda$ .
- Figure 12: Delay  $D$  against arrival rate  $P_{Arr}$  for different values of  $\lambda$  for the unslotted hybrid CDMA/ISMA protocol; Code length = 31; SNR = 20 dB; Number of groups = 8; Group size = 4; a..f : 0.1, 0.5, 1, 2, 5, 10 for  $\lambda$ .
- Figure 13: Throughput  $S$  versus arrival rate  $P_{Arr}$  for comparison between slotted and unslotted hybrid CDMA/ISMA protocol; Code length = 31;  $\lambda=1$ ;  $d=0.01$ ; Number of users = 32. The number of codes used vary from 1 to 32. Slotted(1) means the performance curve with 1 code for the slotted scheme, etc.
- Figure 14: Delay  $D$  versus arrival rate  $P_{Arr}$  for comparison between slotted and unslotted hybrid CDMA/ISMA protocol; Code length = 31,  $\lambda=1$  and  $d=0.01$ ; Number of users = 32; The number of codes used varies from 1 to 32. Slotted(1) means the performance curve with 1 code, etc.

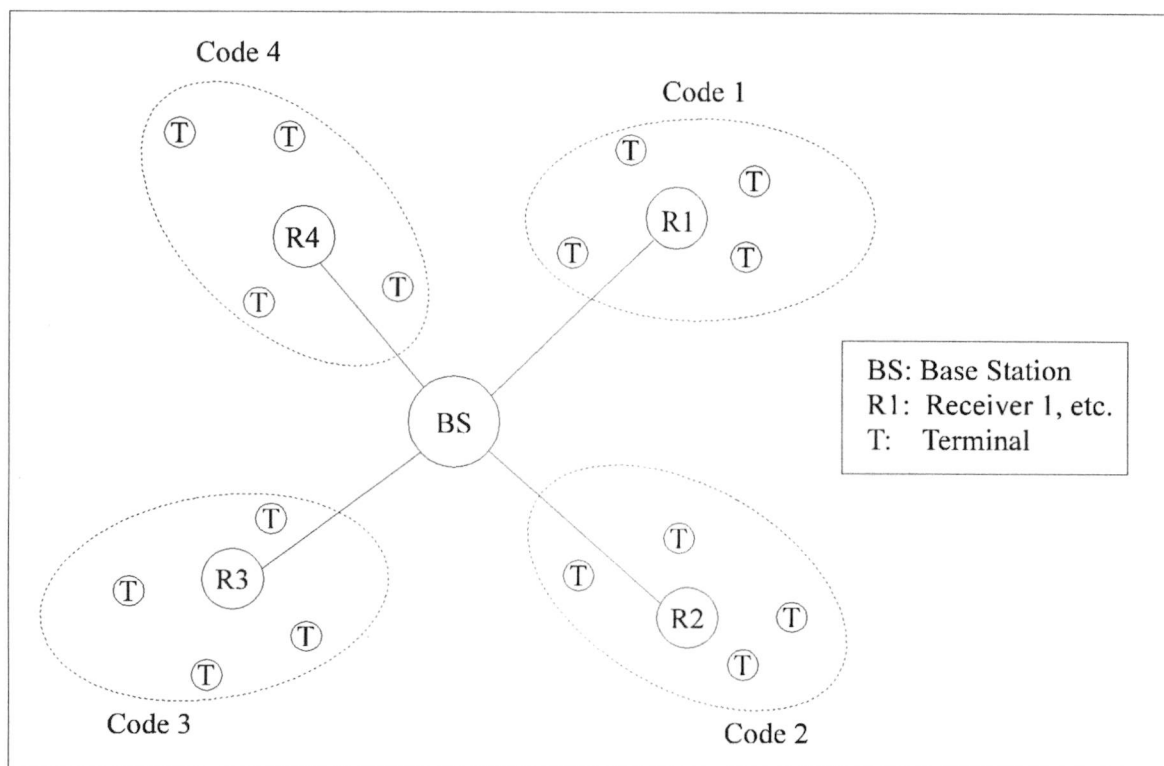


Figure 1: Network structure

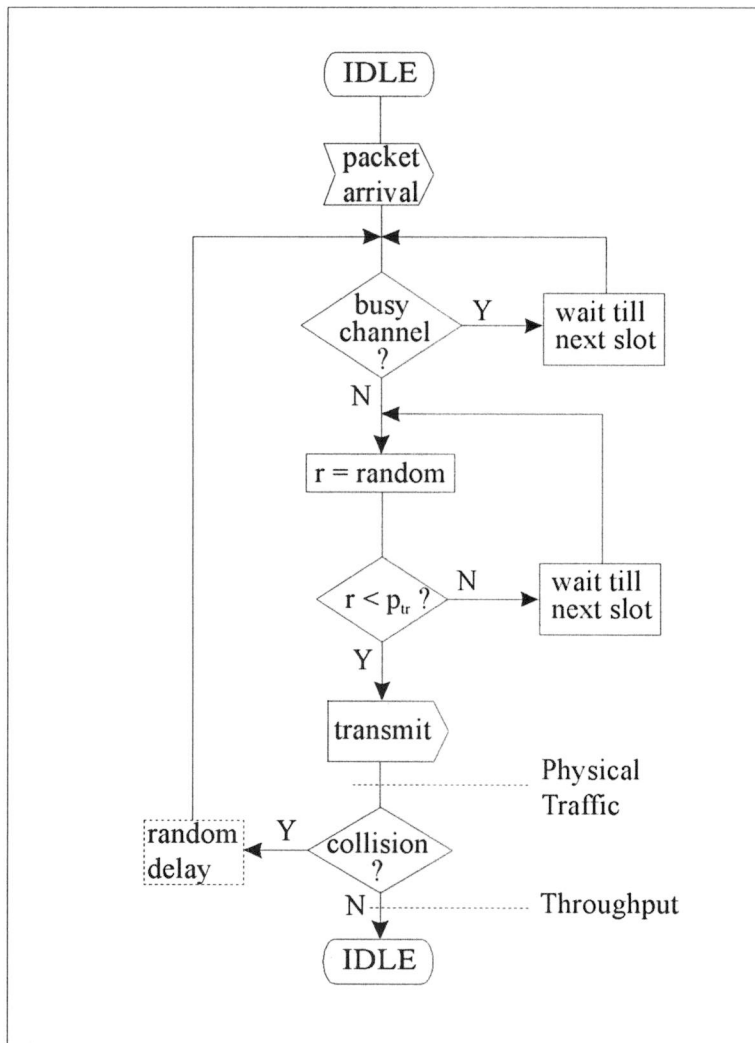


Figure 2: Slotted p-persistent ISMA scheme

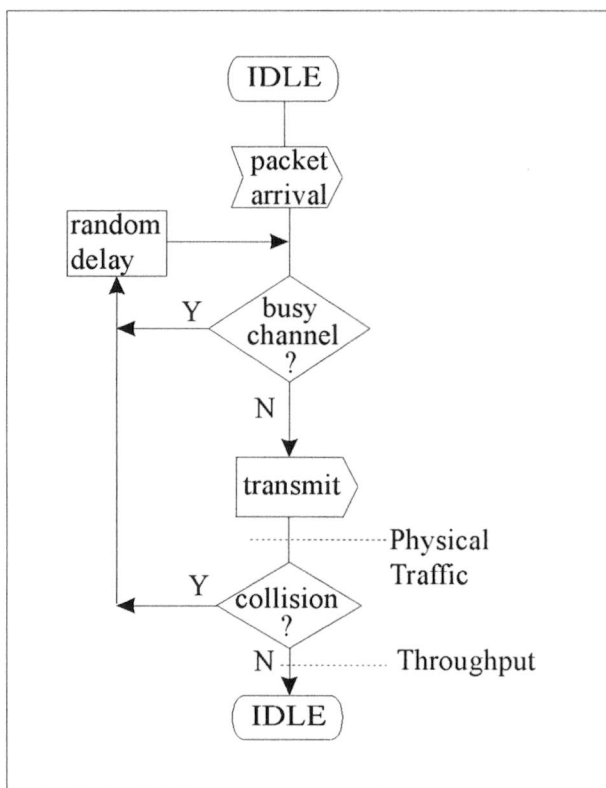


Figure 3: Unslotted non-persistent ISMA scheme



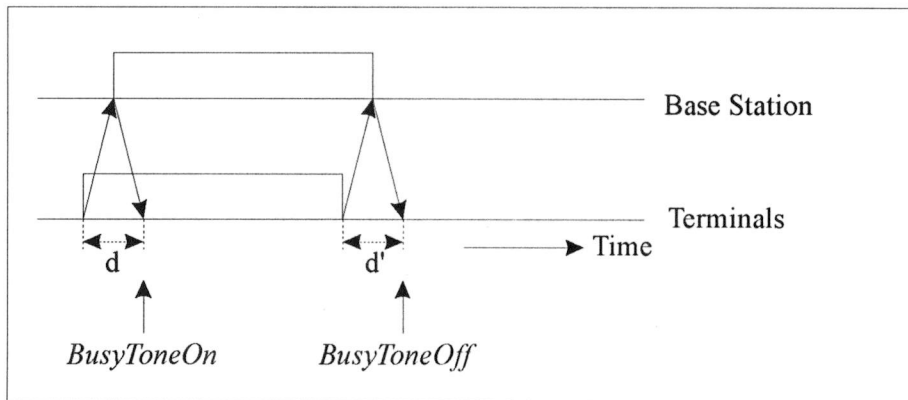


Figure 4: Inhibit delay fraction

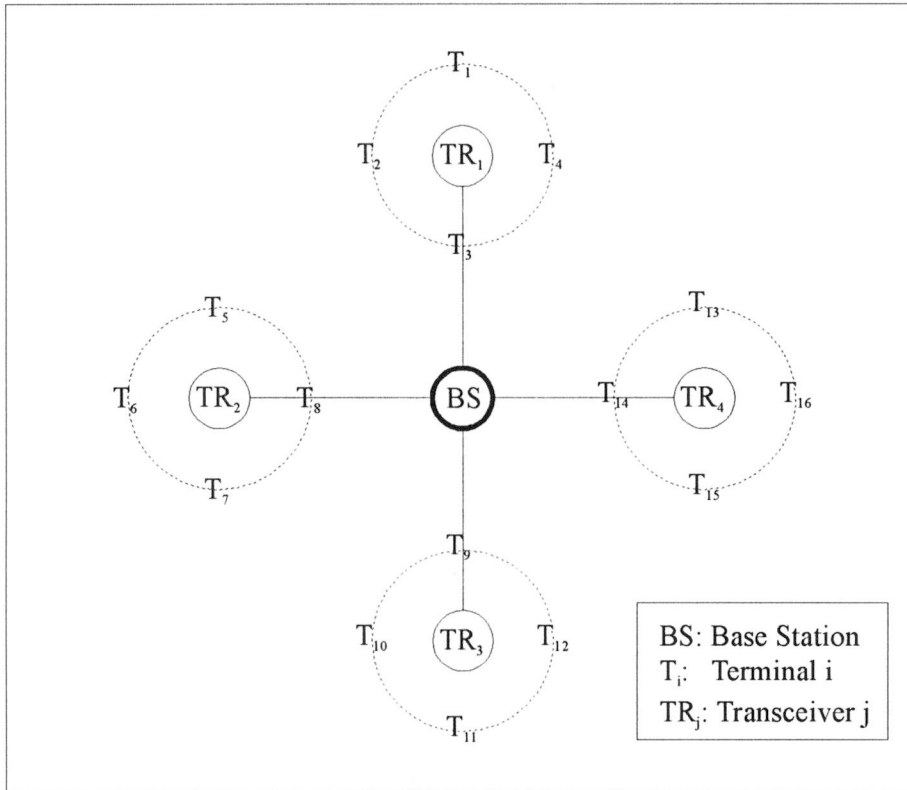


Figure 5: 16-terminal network configuration

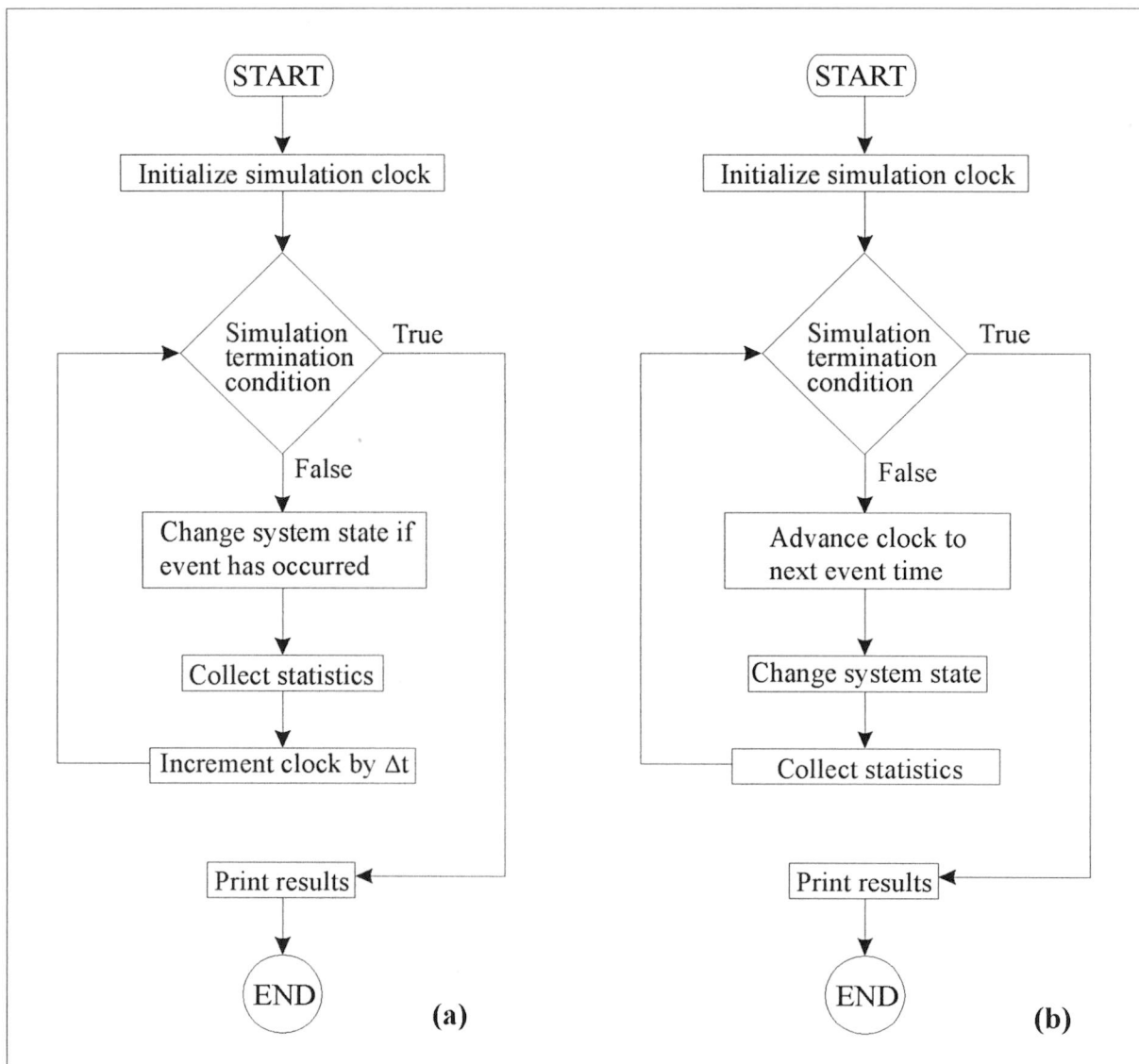


Figure 6: (a) Interval-oriented simulation; (b) Event-oriented simulation

Reset Counters
While <i>Clock</i> < <i>EndTime</i> Do
Advance <i>Clock</i> to next event time
Process Event
Update Events List
Collect Statistics

Figure 7: PSD of main program

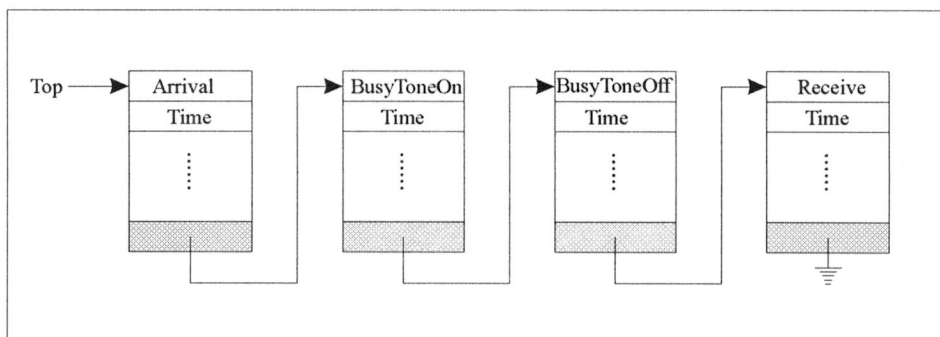


Figure 8: Example of an event list

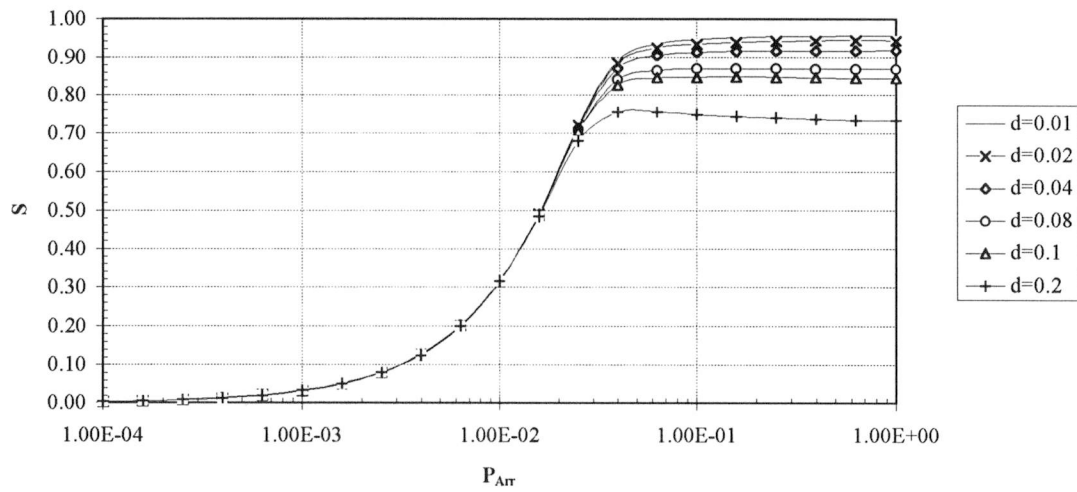


Figure 9: Throughput  $S$  versus arrival rate  $P_{Arr}$  for different values of the inhibit delay fraction  $d$  parameter for the unslotted hybrid CDMA/ISMA protocol; Code length = 31; SNR = 20 dB; Number of groups = 8; Group size = 4.

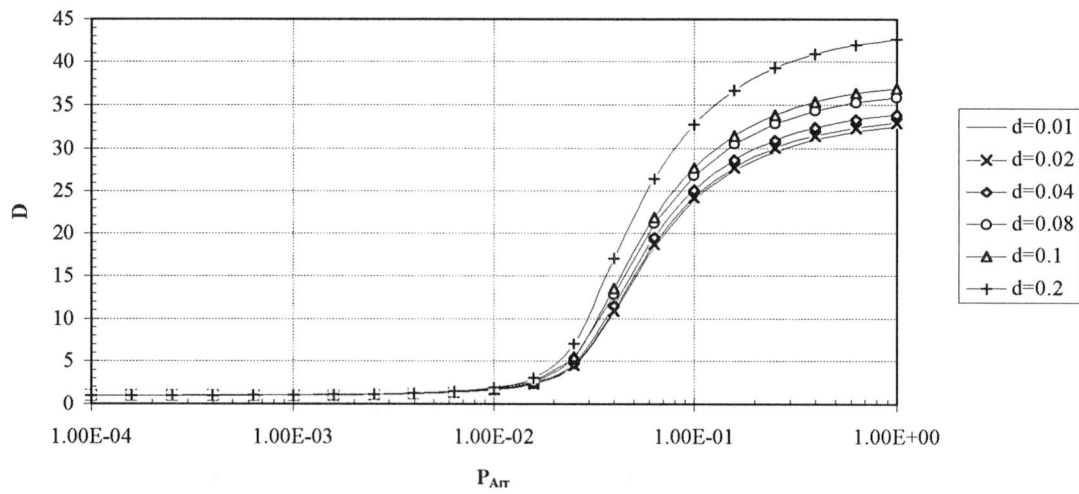


Figure 10:

Delay  $D$  versus arrival rate  $P_{Arr}$  for different values of the inhibit delay fraction  $d$  parameter for the unslotted hybrid CDMA/ISMA protocol; Code length = 31; SNR = 20 dB; Number of groups = 8; Group size = 4.

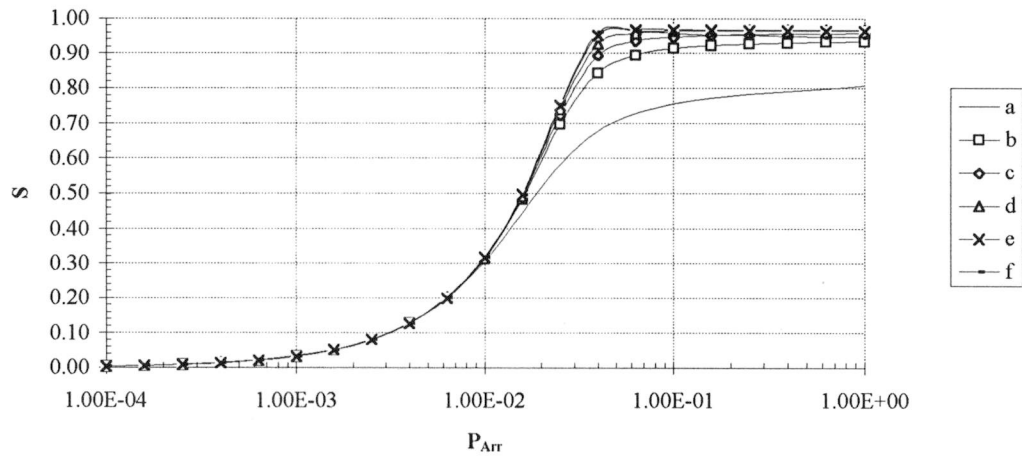


Figure 11: Throughput  $S$  versus arrival rate  $P_{Arr}$  for different values of  $\lambda$  for the unslotted hybrid CDMA/ISMA protocol; Code length = 31; SNR = 20 dB; a..f : 0.1, 0.5, 1, 2, 5, 10 for  $\lambda$ .



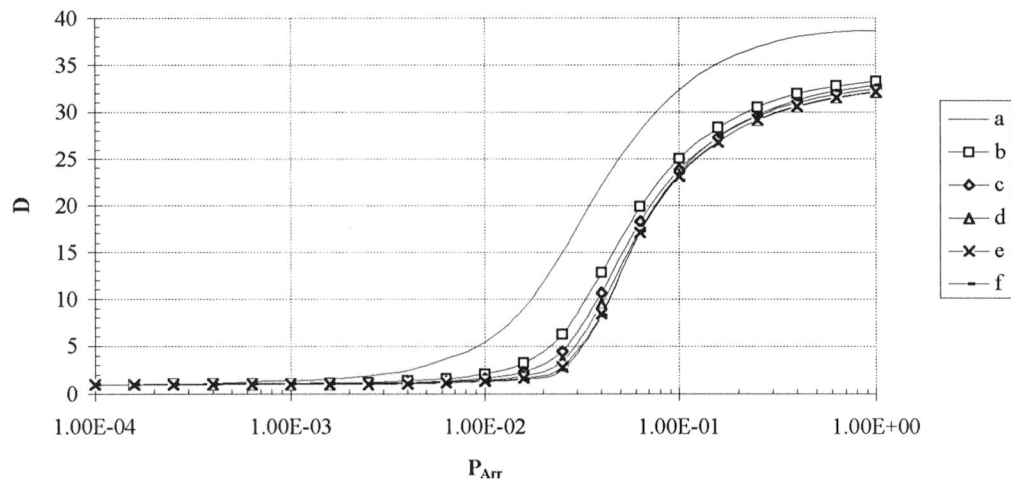


Figure 12: Delay  $D$  against arrival rate  $P_{Arr}$  for different values of  $\lambda$  for the unslotted hybrid CDMA/ISMA protocol; Code length = 31; SNR = 20 dB; Number of groups = 8; Group size = 4; a..f : 0.1, 0.5, 1, 2, 5, 10 for  $\lambda$ .

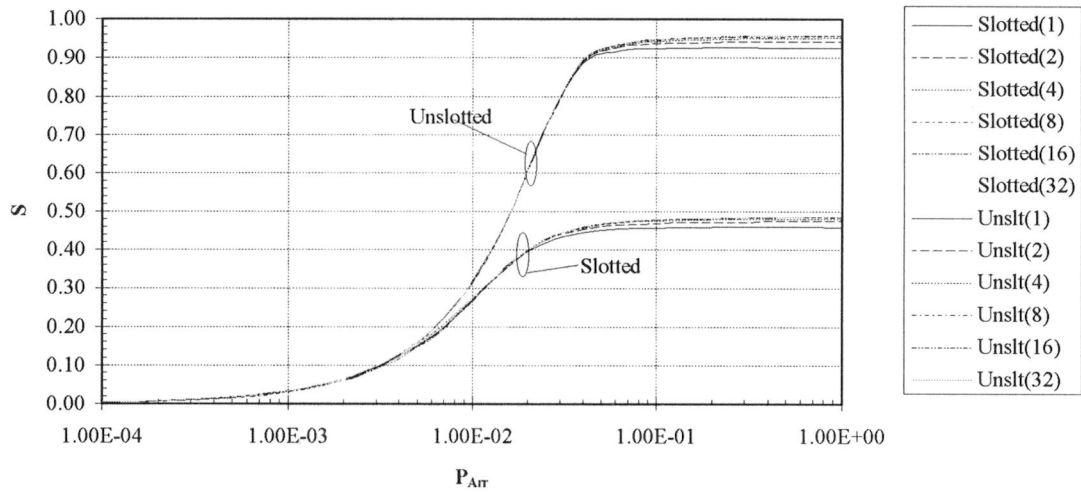


Figure 13: Throughput  $S$  versus arrival rate  $P_{Arr}$  for comparison between slotted and unslotted hybrid CDMA/ISMA protocol; Code length = 31;  $\lambda = 1$ ;  $d = 0.01$ ; Number of users = 32. The number of codes used vary from 1 to 32. Slotted(1) means the performance curve with 1 code for the slotted scheme, etc.

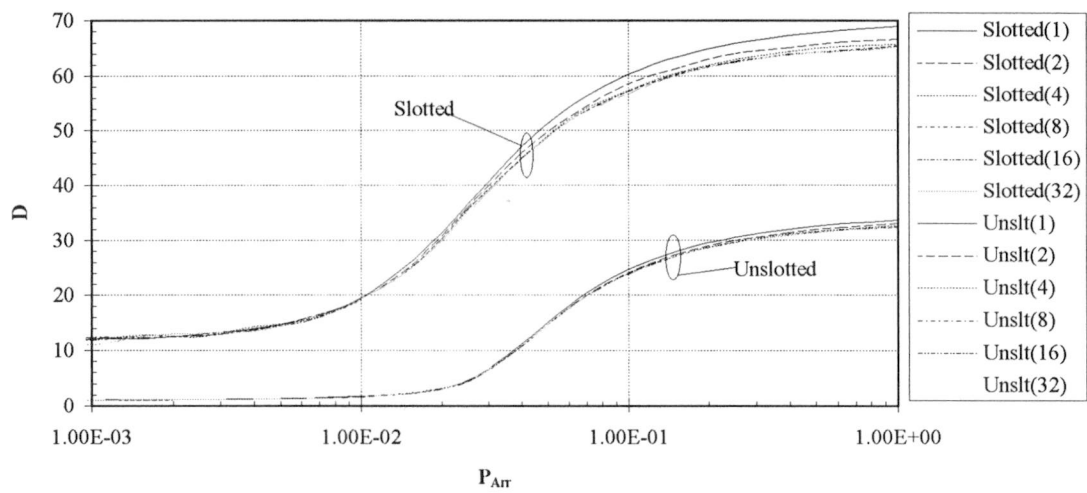


Figure 14: Delay  $D$  versus arrival rate  $P_{Arr}$  for comparison between slotted and unslotted hybrid CDMA/ISMA protocol; Code length = 31,  $\lambda=1$  and  $d=0.01$ ; Number of users = 32; The number of codes used varies from 1 to 32. Slotted(1) means the performance curve with 1 code, etc.

## APPENDIX D:

### Listings of Computer Programs

Three computer programs are included in this Appendix and are listed in the following order:

1. **SimSlotted**

This is a computer simulation program which simulates the behaviour of the slotted hybrid CDMA/ISMA protocol.

2. **SimUnslotted**

This is a computer simulation program which simulates the behaviour of the unslotted hybrid CDMA/ISMA protocol.

3. **CalcUnslotted**

This is a computer program which calculates the performance of the unslotted hybrid CDMA/ISMA protocol.

```

1 {*****}
2 {* SimSlotted is a program designed with the goal to simulate the *}
3 {* 'Slotted Hybrid CDMA/ISMA protocol'. This program calculates the *}
4 {* throughput and delay of a network of terminals by simulation. *}
5 {* *}
6 {* Most of the constants, variables, functions and procedures used here *}
7 {* originates from Huub van Roosmalen. *}
8 {* *}
9 {* The terminals are clustered around distributed receivers at a fixed *}
10 {* distance. The distributed receivers are clustered around the base *}
11 {* station at a fixed distance. The packets from the terminals are *}
12 {* received by the distributed receivers and sent to the base station by *}
13 {* wire or lines. These lines can be seen as primitive concentrators *}
14 {* *}
15 {* Huy Linh Anh Le, August 10 1995 *}
16 {*****}
17 Program SimSlotted(Input, Output);
18
19 Uses Dos, Crt, Ultrax;
20
21 Const MaxTerm = 32; { Maximum number of terminals allowed }
22 PacketLength = 64; { Number of bits per packet }
23
24 Pt = 1; { Transmitter power in Watts }
25 LossExp = 2; { Attenuation parameter for the ... }
26 { ...far field model }
27 c = 299792458; { Speed of light in m/s }
28 f = 1700000000; { Transmitter center frequency }
29 lambda = c/f; { Signal wavelength }
30
31 twoPi = 2 * Pi; { Two times Pi }
32 Pi2 = Pi * Pi; { Pi squared }
33
34 Ptr = 0.99; { Transmission probability }
35 Alpha = 0.1; { The number of Rearrivals per ... }
36 { ...PacketDuration }
37 bitrate = 256*1024; { Bits transmitted per sec. }
38 PacketDuration = 64/bitrate; { Time needed to send a Packet }
39
40 maxcodelength = 127; { Maximum code size }
41 codelength = 31; { Length of code }
42 Tb = 1/bitrate; { Bit duration }
43 Tc = 1/(codelength*bitrate); { Chip duration }
44 Tm = 100E-9; { Maximum delay spread }
45 MaxL = 8; { Maximum number of paths }
46 L = Trunc(Tm/Tc)+1; { Number of paths }
47
48 WarmTime = 21000; { Warm up time in sec. }
49 Interval = 500; { Time between screen updates }
50 EndTime = 125000; { Simulation time in sec. }
51
52 NumSim = 20; { Number of activity levels simulated }
53 LRangeMin = 4; { Terminal activity from 10^-LRangeMin }
54 LRangeMax = 0; { ...to 10^-LRangeMax }
55

```

```

56 Type TermState = (IDLE, BLOCKED, TRANSMIT, INTERFERE);
57 OnOff = (ON, OFF);
58 Matrix = Array[1..MaxTerm, 1..MaxTerm] Of Extended;
59 PosArr = Array[1..MaxTerm] Of Extended;
60 RArr = Array[1..MaxTerm, 1..MaxL] Of Extended;
61 StringArr = Array[1..MaxTerm] Of String;
62 TermArr = Array[1..MaxTerm] Of TermState;
63
64 {-----}
65 p2CRec = ^CRec;
66
67 CRec = Record
68     cc : Array[-maxcodelength..maxcodelength] Of ShortInt;
69     End;
70
71
72 CArr = Array [1..MaxTerm, 1..MaxTerm] Of p2CRec;
73
74 {-----}
75 QueueEntry = RECORD
76     TerminalNo : Integer;
77     Time : Extended;
78     END;
79
80
81 QueueType = RECORD
82     Count,
83     Front,
84     Rear: 0..Maxterm;
85     Entry: ARRAY [1..Maxterm] OF QueueEntry;
86     END;
87
88 {-----}
89 EventType = (Arrival, ReArrival, BusyToneOn, BusyToneOff, Receive);
90
91 ListPointer = ^ListNode;
92 ListNode = RECORD
93     Event : EventType;
94     Time : Extended;
95     TerminalNo : Integer;
96     NextNode : ListPointer
97     END;
98
99 LinkedList = RECORD
100     Head : ListPointer
101     END;
102
103 {-----}
104 Var Count,
105 Front,
106 Rear : Integer; { Counters for QueueType }
107
108 CCat : CArr; { Catalog of code cross correlations }
109
110 PArrH, { Packet Arrival Probability Slotted }

```

```

111   PArrL,           { Packet Arrival Probability Unslotted }
112   Rreceive,        { Receiver radius around base station }
113   Rterm,           { Terminal radius around receivers }
114   SNR              :Extended; { Signal to white Noise Ratio at receiver }
115
116   Nt,              { Number of terminals }
117   GroupSize,        { Number of terminals per group }
118   NumberOfLines :Integer; { Number of groups in the system. This is }
119                     { ... also the number of codes used. }
120
121   DelayArray       :PosArr; { Delay administration for each terminal }
122   DelayCount       :Extended; { Total delay time in Tpd }
123
124   SuccessCount,     { Number of successful packets }
125   ArrivalCount,     { Number of accepted packets }
126   GCount           :LongInt; { Number of collisions occurred }
127
128   S                :Matrix; { Power loss ratio matrix }
129
130   OldClock,         { Clock maintained for screen updates }
131   Clock            :Extended; { Clock maintained by main program }
132
133   EventList        :LinkedList; { Nodes of events linked together }
134   ReceiveQueue     :QueueType; { Queue for received packets }
135
136   BusyTone         :OnOff; { Indicates wheather the channel is free }
137                     { or not. }
138   Terminal         :TermArr; { Terminal state information }
139
140   i                :Integer; { Loopvariable for terminal activity levels }
141
142   ResultFile       :Text; { Result file with original values }
143
144   Throughput,       { Throughput: fraction of the time
145                     ...carrying a successful transmission }
146   D,               { Mean Delay normalized in slots }
147   G                :Extended; { Mean Delay: time between the arrival of
148                     ...the first bit till the receive of the
149                     ...last bit in sec. }
150 {*****}
151 { Power calculates a raised to the power of b }
152 {
153 { Depends on: none
154 { Calls : none
155 { Modifies : none
156 {
157 { Pre : a >= 0 And b >=0
158 { Post: Power = a^b
159 {*****}
160 Function Power(a,b:Extended):Extended;
161 Begin
162   If a<>0
163   Then
164     power := Exp(b*Ln(a))
165   Else
166     If b<>0
167     Then
168       power := 0 { power(0,b) = 0 }
169     Else
170       power := 1; { power(0,0) = 1 }
171 End;{Power}
172
173 {*****}
174 { ShowState shows the terminal states on screen }
175 {
176 { Depends on: config, numberOfLines, terminal
177 { Calls : none
178 { Modifies : none
179 {*****}
180 Procedure ShowState;
181 Var
182   i : Integer;
183   c : Char;
184 Begin
185   GotoXY(12, 22);
186   Write(' 1 2 3 ');
187   GotoXY(12, 23);
188   Write('1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1 2');
189   For i := 1 To Nt Do
190     Begin
191       GotoXY(10+2*i,24);
192       Case Terminal[i] Of
193         IDLE : Write('i');
194         BLOCKED : Write('b');
195         TRANSMIT : Write('t');
196         INTERFERE : Write('f');
197       End;{Case}
198     End;{For i := 1 TO Nt}
199 End;{ShowState}
200
201 {*****}
202 { R calculates a partial cross correlation }
203 {
204 { Depends on: config, CCat, Tc
205 { Calls : none
206 { Modifies : none
207 {*****}
208 Function R(i,j:Integer; tau:Extended):Extended;
209 Var
210   l, { Chip number belonging to tau }
211   cl, { Discrete correlation function for chip l }
212   cl1 :Integer; { Discrete correlation function for chip l+1 }
213 Begin
214   l := Trunc(tau/Tc);
215   cl1 := CCat[i,j]^cc[l+1-codelength];
216   If l >= 1
217   Then
218     Begin
219       cl := CCat[i,j]^cc[l-codelength];
220       R := cl*Tc+(cl1-cl)*(tau-l*Tc);

```

```

221 End
222 Else
223   R := tau*Cl1;
224 End;{R}
225
226 {*****}
227 { Rhat calculates a partial cross correlation }
228 { }
229 { Depends on: config, CCat, Tc }
230 { Calls : none }
231 { Modifies : none }
232 {*****}
233 Function Rhat(i,j:Integer; tau:Extended):Extended;
234 Var
235   l, { Chip number belonging to tau }
236   Cl, { Discrete correlation function for chip l }
237   Cl1 :Integer; { Discrete correlation function for chip l+1 }
238 Begin
239   l := Trunc(tau/Tc);
240   Cl := CCat[i,j].cc[l];
241   If l < codelength-1
242   Then
243     Begin
244       Cl1 := CCat[i,j].cc[l+1];
245       Rhat := Cl*Tc+(Cl1-Cl)*(tau-l*Tc);
246     End
247   Else
248     Rhat := ((l+1)*Tc-tau)*Cl;
249 End;{Rhat}
250
251 {*****}
252 { InitCCat sets up a catalog of discrete aperiodic cross correlation }
253 { functions for a set of (Gold) codes passed by gc. The catalog consists of }
254 { a two dimensional array of pointers. These pointers refer to }
255 { one-dimensional arrays containing the cross correlation information. This }
256 { information is in essence the number of similarities minus the numbers }
257 { of dissimilarities between two codes for a given discrete phase shift }
258 { }
259 { Depends on: config, p2CRec }
260 { Calls : none }
261 { Modifies : none }
262 {*****}
263 Procedure InitCCat(Var CCat:CArr; gc:StringArr);
264 Var
265   i, { Loop variable }
266   j, { Loop variable }
267   k, { Loop variable }
268   l, { Discrete phase shift }
269   t :Integer; { Correlation result for shift l }
270 Begin
271   Write('Calculating ', Nt, ' by ', Nt, ' correlations');
272   For i := 1 To Nt Do { Double loop over all codes }
273     For j := 1 To Nt Do
274       Begin
275         GotoXY(70,1); { Show which codes are being done }

```

```

276   Write(i:4,j:4);
277   CCat[i,j] := New(p2CRec); { Allocate space for new array }
278
279   { Calculate cross-correlation }
280   For l := -codelength To codelength Do
281     Begin
282       t := 0; { Reset correlation value }
283       If l < 0 { Different formulas for pos and }
284       Then { neg phase shifts }
285         For k := 0 To codelength-1+l Do
286           If gc[i][1+((k-l) Mod codelength)] = gc[j][k+1]
287           Then
288             Inc(t)
289           Else
290             Dec(t)
291         Else
292           For k := 0 To codelength-l-1 Do
293             If gc[i][1+(k Mod codelength)] =
294             gc[j][1+((k+l) Mod codelength)]
295             Then
296               Inc(t)
297             Else
298               Dec(t);
299           CCat[i,j].cc[l] := t;
300         End;
301       End;
302     End;{InitCCat}
303
304 {*****}
305 { DisposeCCat frees memory space taken up by the correlation catalog }
306 { }
307 { Depends on: config }
308 { Calls : none }
309 { Modifies : none }
310 {*****}
311 Procedure DisposeCCat(CCat:CArr);
312 Var
313   i, { Loop variable }
314   j :Integer; { Loop variable }
315 Begin
316   For i := 1 To Nt Do { Loop over all catalog entries }
317     For j := 1 To Nt Do
318       Dispose(CCat[i,j]); { Free memory of element i,j }
319     End;{DisposeCCat}
320
321 {*****}
322 { InitTerm initializes all terminals as free }
323 { }
324 { Depends on: config }
325 { Calls : none }
326 { Modifies : pfn }
327 {*****}
328 Procedure InitTerm(Var Terminal:TermArr);
329 Var

```

```

331 i :Integer; { Loop variable }
332 Begin
333   For i:= 1 To Nt Do
334     terminal[i] := IDLE;
335 End;{InitTerm}
336
337 {*****}
338 PROCEDURE InitDelayArray(VAR DelayArray:PosArr);
339 VAR
340   i : Integer; { Loop variable }
341 BEGIN
342   FOR i := 1 TO MaxTerm DO
343     DelayArray[i] := 0;
344 END;{InitDelayArray}
345
346 {*****}
347 PROCEDURE InitBusyTone(VAR BusyTone:OnOff);
348 BEGIN
349   BusyTone := Off;
350 END;{InitBusyTone}
351
352 {*****}
353 { Initcode reads Gold codes from a file and has the cross correlation }
354 { catalog setup }
355 {
356 { Depends on: config }
357 { Calls : initCCat }
358 { Modifies : pfn }
359 {*****}
360 Procedure InitCodes(VAR CCat:CArr; groupSize, numCodes:Integer);
361 Var
362   i, { Loop variable }
363   j :Integer; { Loop variable }
364   f :text; { File containing the Gold codes }
365   codeCount, { Length of transmission code }
366   cfile :String; { Codefile name }
367   gc :StringArr; { Gold codes as read from file }
368 Begin
369   ClrScr;
370   Str(codelength, codeCount); { Assign appropriate codefile name }
371   cfile := 'codes\' + codeCount + 'chip.dat';
372   Assign(f, cfile); { Open code file }
373   Reset(f);
374   For i := 1 To Nt Do { Read the codes from file }
375     ReadLn(f,gc[i]);
376   Close(f);
377   If numCodes <> Nt { Reassign codes according to reuse scheme }
378     Then { e.g. 12345678 -> 11223344 for 2 term/grp }
379     For i := numCodes DownTo 1 Do
380       For j := groupSize DownTo 1 Do
381         gc[(i-1)*groupSize+j] := gc[i];
382       initCCat(CCat, gc);
383     { Initialize catalog of cross correlations }
384 End;{InitCodes}
385

```

```

386 {*****}
387 { InitPathLoss sets up a table of transmitter to receiver path gains from }
388 { all transmitters to all receivers. All gains are between 0 and 1. When }
389 { groups are larger than one terminal duplicate entries will exist in the }
390 { table because there are less receivers than terminals then. }
391 {
392 { Depends on: config, lambda }
393 { Calls : Power }
394 { Modifies : none }
395 {*****}
396 Procedure InitPathGain(VAR S:Matrix; groupSize, numberOfLines :Integer;
397   Rreceive, Rterm, lossexp:Extended );
398 Var
399   gamma, { Angle of receiver relative to x-axis }
400   theta :Extended; { Angle of transmitter relative to x-axis }
401   xt, yt, { Transmitter coordinates }
402   xr, yr :PosArr; { Receiver coordinates }
403   i, { Loop variable }
404   j, { Loop variable }
405   k :Integer; { Index variable }
406 Begin
407   For i := 1 To numberOfLines Do { Loop over all receivers }
408     For j := 1 To groupSize Do { Loop over all group terminals }
409       Begin
410         k := (i-1) * groupSize + j; {Determine index current term. }
411         gamma := 2*Pi * (i-1)/numberOfLines; {Calc. receiver angle }
412         xr[k] := Rreceive * Cos(gamma); {Calc. receiver x-coord. }
413         yr[k] := Rreceive * Sin(gamma); {Calc. receiver y-coord. }
414         theta := 2*Pi*(j-1)/groupSize+gamma+Pi; {Calc. transmitter angle }
415         xt[k] := xr[k] + Rterm * Cos(theta); {Calc. transmitter x-coord.}
416         yt[k] := yr[k] + Rterm * Sin(theta); {Calc. transmitter y-coord.}
417       End;
418       For i := 1 To Nt Do {Double loop over all terminals}
419         For j := 1 To Nt Do
420           Begin
421             { Calculate path gains according to far field model }
422             S[i,j] := Power(lambda/(4*Pi*Sqr(Sqr(xt[i]-xr[j])
423               +Sqr(yt[i]-yr[j]))),lossexp);
424           End
425 End;{InitPathGain}
426
427 {*****}
428 { Rayleigh randomly draws a value according to the rayleigh distribution }
429 {
430 { Depends on: none }
431 { Calls : duni }
432 { Modifies : pfn }
433 {*****}
434 Function Rayleigh(ave:Extended):Extended;
435 Var
436   r,
437   u,
438   v,
439   x,
440   y :Extended;

```



```

441 Begin
442   Repeat
443     u := 2*duni-1;
444     v := 2*duni-1;
445     r := sqrt(u)+sqrt(v);
446   Until (r<1.0) and (r>0.0);
447   r := Sqrt(-2*Ln(r)/r);
448   x := u*r;
449   y := v*r;
450   Rayleigh:=Sqrt(Sqr(x)+Sqr(y))*ave*Sqrt(2/Pi);
451 End;
452
453 {*****}
454 { MarcumQ calculates the Marcum Q-function }
455 { }
456 { Depends on: none }
457 { Calls : none }
458 { Modifies : pfn }
459 {*****}
460 Function MarcumQ(a,b:Extended):Extended;
461 Var
462   aa,
463   ab,
464   ac,
465   bc,
466   bb,
467   ba,
468   dd,
469   d1 :Extended;
470   n :Integer;
471 Begin
472   If a = 0
473   Then
474     If b < 30
475     Then
476       MarcumQ := Exp(-Sqr(b)/2)
477     Else
478       MarcumQ := 0
479   Else
480   Begin
481     If a < b
482     Then
483       Begin
484         ab := 1;
485         dd := a/b;
486       End
487     Else
488       Begin
489         ab := 0;
490         dd := b/a;
491       End;
492     aa := 0 ;
493     bb := 0.5;
494     ba := 0 ;
495     d1 := dd;

```

```

496   n := 0;
497   Repeat
498     n := n + 1;
499     ac := d1+2*n*ab/(a*b)+aa;
500     bc := 1+2*n*bb/(a*b)+ba;
501     d1 := dd*d1;
502     aa := ab;
503     ab := ac;
504     ba := bb;
505     bb := bc;
506   Until bc > 1e12;
507   If a < b
508   Then
509     MarcumQ := (ac/(2*bc))*Exp(-Sqr(a-b)/2)
510   Else
511     MarcumQ := 1-(ac/(2*bc))*Exp(-Sqr(a-b)/2);
512   End;
513 End;
514
515 {*****}
516 { NegExpI0 approximates the modified Bessel function of the first kind }
517 { using the recurrence relation  $I_0(z) = I_0(z) + 2n/z I_1(z)$  }
518 {  $I_0, n-1$   $I_0, n+1$   $I_0, n$  }
519 { }
520 { and then multiplies the result with  $\text{Exp}(-(a^2+b^2)/2)$ . }
521 { }
522 { Depends on: none }
523 { Calls : none }
524 { Modifies : pfn }
525 {*****}
526 Function NegExpI0(a,b:Extended):Extended;
527 Var
528   n :Integer; { Loop variable }
529   x,
530   Ja, {  $I_0, n-1$  }
531   Jb, {  $I_0, n$  }
532   Jc, {  $I_0, n+1$  }
533   tmp,
534   t :Extended; { Intermediate result }
535 Begin
536   tmp := (Sqr(a)+Sqr(b))/2;
537   x := a*b;
538   If x = 0
539   Then
540     If tmp < 150
541     Then
542       NegExpI0 := Exp(-tmp)/2
543     Else
544       negExpI0 := 0
545   Else
546   Begin
547     Jc := 0;
548     Jb := 1;
549     t := 0;
550     For n := 20 DownTo 1 Do

```

```

551      Begin
552          Ja := (2*n/x)*Jb+Jc;
553          Jc := Jb;
554          Jb := Ja;
555          t := t + Ja;
556      End;
557      NegExpI0 := Exp(x-(Sqr(a)+Sqr(b))/2)*Ja/(2*t-Ja);
558  End;
559 End;
560
561 {*****}
562 { CalcMu1 calculates the Mu1 parameter for the bit error formula in }
563 { Rayleigh fading environments }
564 { }
565 { Depends on: L }
566 { Calls : R, Rhat }
567 { Modifies : none }
568 {*****}
569 Function CalcMu1(u, h:Integer; tau:RArr; mu2:Extended):Extended;
570 Var
571     tmp :Extended; { Intermediate result }
572     i :Integer; { Loop variable }
573 Begin
574     tmp := 0;
575     For i := 1 To L Do
576         If i <> h
577             Then
578                 tmp := tmp + R(u,u,tau[u,i])*Rhat(u,u,tau[u,i])*s[u,u];
579
580     CalcMu1 := 4*2*Pt*tmp + mu2
581 End;
582
583 {*****}
584 { CalcMu2 calculates the Mu2 parameter for the bit error formula in }
585 { Rayleigh fading environments }
586 { }
587 { Depends on: config, L, Pt, terminal }
588 { Calls : R, Rhat }
589 { Modifies : none }
590 {*****}
591 Function CalcMu2(u, h:Integer; tau:RArr; sigma_n2:Extended):Extended;
592 Var
593     tmp :Extended; { Intermediate result }
594     i, { Loop variable }
595     j :Integer; { Loop variable }
596 Begin
597     tmp := 0;
598     For i := 1 To Nt Do
599         If (terminal[i] = INTERFERE) And (i <> u)
600             Then
601                 For j := 1 To L Do
602                     tmp := tmp + (Sqr(R(u,i,tau[i,j]))
603                         + Sqr(Rhat(u,i,tau[i,j]))) * s[i,u];
604
605     For j := 1 To L Do

```

```

606         If j <> h
607             Then
608                 tmp := tmp + (Sqr(R(u,u,tau[u,j]))
609                     + Sqr(Rhat(u,u,tau[u,j]))) * s[u,u];
610
611     CalcMu2 := 2*(2*Pt*tmp+sigma_n2);
612 End;
613
614 {*****}
615 { CalcMu12 calculates the Mu12 parameter for the bit error formula in }
616 { Rayleigh fading environments }
617 { }
618 { Depends on: config, L, Pt, terminal }
619 { Calls : R, Rhat }
620 { Modifies : none }
621 {*****}
622 Function CalcMu12(u,h:Integer;tau:RArr):Extended;
623 Var
624     tmp :Extended; { Intermediate result }
625     i, { Loop variable }
626     j :Integer; { Loop variable }
627 Begin
628     tmp := 0;
629     For i := 1 To Nt Do
630         If (terminal[i] = INTERFERE) And (i <> u)
631             Then
632                 For j := 1 To L Do
633                     tmp := tmp + R(u,i,tau[i,j])*Rhat(u,i,tau[i,j])*s[i,u];
634
635     For j := 1 To L Do
636         If j <> h
637             Then
638                 tmp := tmp + (Sqr(Rhat(u,u,tau[u,j]))
639                     + R(u,u,tau[u,j])*Rhat(u,u,tau[u,j])) * S[u,u];
640
641     CalcMu12 := 2*2*Pt*tmp;
642 End;
643
644 {*****}
645 { CalculatePbe computes the bit error probability for a terminal, given }
646 { which other terminals transmit }
647 { }
648 { Depends on: config, L, Pt, S, Tb, terminal }
649 { Calls : duni, I0, MarcumQ, Rayleigh }
650 { Modifies : none }
651 {*****}
652 Function CalculatePbe(u:Integer):Extended;
653 Var
654     mu1,
655     mu2,
656     mu12,
657     a,
658     b,
659     bmax, { Maximum path gain }
660     tmp,

```

```

661 m,
662 mq,
663 mb,
664 sigma_n2:Extended;      { Noise power }
665 beta,                    { Rayleigh distributed path gains }
666 tau      :RArr;          { Uniformly [0,Tb] distributed path delays }
667 h,                      { Number of path with largest gain }
668 i,                      { Loop variable }
669 j          :Integer;      { Loop variable }
670 Begin
671   { Assign random delays and path gains to transmitting terminals }
672   For i := 1 To Nt Do
673     If terminal[i] = INTERFERE
674     Then
675       For j := 1 To L Do
676         Begin
677           tau[i,j] := Tb*duni;
678           beta[i,j] := Rayleigh(Sqrt(S[i,u]));
679         End
680       Else
681         For j := 1 To L Do
682           Begin
683             tau[i,j] := 0;
684             beta[i,j] := 0;
685           End;
686
687   { Determine what the largest gain is }
688   bmax := 0;
689   h := 1;
690   For i := 1 To L Do
691     If beta[u,i]>bmax
692     Then
693       Begin
694         bmax := beta[u,i];
695         h := i;
696       End;
697
698   { Determine noise power relative to largest incoming signal }
699   sigma_n2 := power(10, -snr/10)*Sqr(bmax)*Pt*Tb;
700
701   { Calculate bit error probability parameters }
702   mu2 := CalcMu2(u, h, tau, sigma_n2);
703   mu1 := CalcMu1(u, h, tau, mu2);
704   mu12 := CalcMu12(u, h, tau);
705   m := Sqrt(2*Pt)*bmax*Tb;
706   a := m*Abs(1/Sqrt(mu1)-1/Sqrt(mu2))/Sqrt(2);
707   b := m*(1/Sqrt(mu1)+1/Sqrt(mu2))/Sqrt(2);
708   mq := MarcumQ(a,b);
709   mb := NegExpI0(a,b);
710
711   { Calculate the bit error probability }
712   tmp := ( (1 + mu12/Sqrt(mu1*mu2))*mb )/2;
713   CalculatePbe := mq-tmp
714 End;
715

```

```

716 {*****}
717 {      Q U E U E S      }
718 {*****}
719
720 {*****}
721 { CreateQueue }
722 { Pre : None }
723 { Post: The queue Q has been created and is initialized to be empty. }
724 {*****}
725
726 PROCEDURE CreateQueue(VAR Q:QueueType);
727 BEGIN
728   WITH Q DO
729     BEGIN
730       Count := 0;
731       Front := 1;
732       Rear := 0
733     END
734 END;{CreateQueue}
735
736 {*****}
737 PROCEDURE Append (TerminalNo:Integer; Time:Extended; VAR Q:QueueType);
738 VAR
739   X : QueueEntry;
740 BEGIN
741   WITH Q DO
742     IF Count = MaxTerm THEN
743       WRITELN ('Error: Attempt to append an entry to a full queue')
744     ELSE
745       BEGIN
746         X.TerminalNo := TerminalNo;
747         X.Time := Time;
748         Count := Count + 1;
749         Rear := (Rear MOD MaxTerm) + 1;
750         Entry [Rear] := X;
751       END
752     END;{Append}
753
754 {*****}
755 PROCEDURE Remove (VAR X: QueueEntry; VAR Q:QueueType);
756 BEGIN
757   WITH Q DO
758     IF Count = 0 THEN
759       WRITELN ('Error: Attempt to remove an entry from an empty queue')
760     ELSE
761       BEGIN
762         Count := Count - 1;
763         X := Entry [Front];
764         Front := (Front MOD MaxTerm) + 1
765       END
766     END;{Remove}
767
768 {*****}
769 FUNCTION QueueSize (VAR Q: QueueType): INTEGER;
770 BEGIN

```

```

771 QueueSize := Q.Count
772 END;{QueueSize}
773
774 {*****}
775 FUNCTION QueueEmpty (VAR Q: QueueType): BOOLEAN;
776 BEGIN
777     QueueEmpty := (Q.Count = 0)
778 END;{QueueEmpty}
779
780 {*****}
781 FUNCTION QueueFull (VAR Q: QueueType): BOOLEAN;
782 BEGIN
783     QueueFull := (Q.Count = MaxTerm)
784 END;{QueueFull}
785
786 {*****}
787 PROCEDURE ShowQueue (VAR Q : QueueType);
788 VAR
789     x, y : INTEGER;
790     x1 : QueueEntry;
791 BEGIN
792     {WRITELN('De inhoud van de queue is:');}
793     x := QueueSize (Q);
794     WITH Q DO
795         y := front;
796         WHILE x > 0 DO
797             BEGIN
798                 x1 := Q.Entry[y];
799                 WRITELN('TN: ',x1.TerminalNo,'Time: ',x1.Time);
800                 x := x-1;
801                 y := (y MOD MaxTerm) + 1
802             END;
803             WRITELN;
804         END;{ShowQueue}
805
806 {*****}
807 {
808     L I N K E D - L I S T
809 }
810 {*****}
811
812 {*****}
813 {* CreateList initializes linked list LL to be empty *}
814 {* *}
815 {* Pre: None *}
816 {* Post: The list LL has been created and is initialized to be empty *}
817 {*****}
818 PROCEDURE CreateList (VAR LL: LinkedList);
819 BEGIN
820     LL.head := NIL
821 END;{CreateList}
822
823 {*****}
824 PROCEDURE InsertEvent (NewNode:ListPointer; T:Extended; VAR LL:LinkedList);
825 VAR

```

```

826 Current,
827 Trailing : ListPointer;
828 BEGIN
829     Current := LL.head;
830     Trailing := NIL;
831     WHILE (Current^.Time <= T) AND (Current^.NextNode <> NIL) DO
832         BEGIN
833             Trailing := Current;
834             Current := Current^.NextNode;
835         END;
836     IF (Current^.time <= T) AND (Current^.NextNode = NIL)
837     THEN
838         BEGIN
839             Current^.NextNode := NewNode;
840             NewNode^.NextNode := NIL;
841         END
842     ELSE IF (Current^.Time > T) AND (Trailing = NIL)
843     THEN
844         BEGIN
845             NewNode^.NextNode := LL.head;
846             LL.head := NewNode;
847         END
848     ELSE
849         BEGIN
850             NewNode^.NextNode := Current;
851             Trailing^.NextNode := NewNode;
852         END;
853     END;{InsertEvent}
854
855 {*****}
856 PROCEDURE CreateEvent (E:EventType; T:Extended; TN:Integer; VAR LL:LinkedList);
857
858 VAR
859     NewNode : ListPointer;
860 BEGIN
861     New(NewNode);
862     NewNode^.Event := E;
863     NewNode^.Time := T;
864     NewNode^.TerminalNo := TN;
865     IF LL.head = NIL
866     THEN
867         BEGIN
868             NewNode^.NextNode := NIL;
869             LL.head := NewNode;
870         END
871     ELSE
872         BEGIN
873             InsertEvent (NewNode, T, LL);
874         END;
875     END;{CreateEvent}
876
877 {*****}
878 PROCEDURE ShowList (VAR LL:LinkedList);
879 VAR
880     Current : ListPointer;

```

```

881 i      : Integer;
882 Event  : String;
883 BEGIN
884     i      := 1;
885     Current := LL.head;
886     ClrScr;
887     WriteLn('The list consists of: (Press ENTER for the nodes to show up)');
888     WriteLn;
889     IF Current = NIL THEN
890         BEGIN
891             WRITELN('EventList is empty! (WriteList)');
892         END
893     ELSE
894         WHILE (Current <> NIL) DO
895             BEGIN
896                 CASE Current^.Event OF
897                     Arrival   : Event := 'Arrival';
898                     ReArrival : Event := 'ReArrival';
899                     BusyToneOn : Event := 'BusyToneOn';
900                     BusyToneOff : Event := 'BusyToneOff';
901                     Receive   : Event := 'Receive';
902                 ELSE WRITELN('ERROR in EventList (ShowList)!!!!');
903             END;
904             WRITELN('Node:', i, '      Event:', Event, 'Time:',
905                 Current^.Time, '      Terminal:', Current^.TerminalNo);
906             i := i + 1;
907             Current := Current^.NextNode;
908             ReadLn;
909             END;
910             WriteLn('End of list');
911             ReadLn;
912 END; {ShowList}
913
914 {*****}
915 PROCEDURE ProcessArrival(FirstNode:ListPointer; VAR BusyTone:OnOff;
916     VAR Terminal:TermArr; VAR LL:LinkedList;
917     VAR Q:QueueType);
918 VAR
919     RandomDelay,
920     Time,
921     TInterArr : Extended;
922     TN        : Integer;
923 BEGIN
924     Clock      := FirstNode^.Time;
925     TN         := FirstNode^.TerminalNo;
926     Time       := FirstNode^.Time;
927     TInterArr := Trunc(-(ln(1-DUNI))/PArrH)+1;
928     CreateEvent(Arrival, Time + TInterArr, TN, LL);
929     IF Terminal[TN] = IDLE
930     THEN
931         BEGIN
932             Terminal[TN] := BLOCKED;
933             Inc(ArrivalCount);
934             IF (DUNI > Ptr)
935             THEN

```

```

936         BEGIN
937             CreateEvent(ReArrival, Time + 1, TN, LL);
938             DelayArray[TN] := DelayArray[TN] + 1;
939             Inc(GCount);
940         END
941     ELSE
942         BEGIN
943             Terminal[TN] := TRANSMIT;
944             Append(TN, Time, Q);
945             CreateEvent(Receive, Time+ 0.95, TN, LL);
946             DelayArray[TN] := DelayArray[TN] + 1;
947         END
948     END
949 END; {ProcessArrival}
950
951 {*****}
952 PROCEDURE ProcessReArrival(FirstNode:ListPointer; VAR BusyTone:OnOff;
953     VAR Terminal:TermArr; VAR LL:LinkedList;
954     VAR Q:QueueType);
955 VAR
956     RandomDelay,
957     Time,
958     TInterArr : Extended;
959     TN        : Integer;
960 BEGIN
961     Clock      := FirstNode^.Time;
962     TN         := FirstNode^.TerminalNo;
963     Time       := FirstNode^.Time;
964     IF DUNI > Ptr
965     THEN
966         BEGIN
967             CreateEvent (ReArrival, Time + 1, TN, LL);
968             DelayArray[TN] := DelayArray[TN] + 1;
969             Inc(GCount);
970         END
971     ELSE
972         BEGIN
973             Terminal[TN] := TRANSMIT;
974             Append(TN, Time, Q);
975             CreateEvent(Receive, Time+0.95, TN, LL);
976             DelayArray[TN] := DelayArray[TN] + 1;
977         END
978     END; {ProcessReArrival}
979
980 {*****}
981 PROCEDURE ProcessReceive(FirstNode:ListPointer; VAR BusyTone:OnOff;
982     VAR Terminal:TermArr; VAR LL:LinkedList;
983     VAR Q:QueueType);
984 VAR
985     X2          : QueueEntry;
986
987     TN,
988     TN2         : Integer;
989
990     RandomDelay,

```

```

991 Time,
992 Time2,
993 Pbe,
994 Psp      : Extended;
995
996 BEGIN
997   Clock := FirstNode^.Time+0.05;
998   Time  := FirstNode^.Time+0.05;
999   IF QueueEmpty(Q) = TRUE
1000   THEN
1001     BEGIN
1002       TN      := FirstNode^.TerminalNo;
1003       RandomDelay := Trunc(-(ln(1-DUNI))/Alpha)+1;
1004       Terminal[TN] := BLOCKED;
1005       CreateEvent(ReArrival, Time+RandomDelay, TN, LL);
1006       DelayArray[TN] := DelayArray[TN] + RandomDelay;
1007       Inc(GCount);
1008     END
1009   ELSE
1010     BEGIN
1011       TN      := FirstNode^.TerminalNo;
1012       Terminal[TN] := INTERFERE;
1013       WHILE NOT QueueEmpty(Q) DO
1014         BEGIN
1015           Remove(X2, Q);
1016           TN2 := X2.TerminalNo;
1017           Terminal[TN2] := INTERFERE;
1018         END;
1019       Pbe := CalculatePbe(TN);
1020       Psp := Power(1-Pbe, PacketLength);
1021     END
1022   IF DUNI < Psp
1023   THEN
1024     BEGIN
1025       Inc(SuccessCount);
1026       Terminal[TN] := IDLE;
1027       DelayCount := DelayCount + DelayArray[TN] + 0.5;
1028       DelayArray[TN] := 0;
1029     END
1030   ELSE
1031     BEGIN
1032       Terminal[TN] := BLOCKED;
1033       RandomDelay := Trunc(-(ln(1-DUNI))/Alpha)+1;
1034       CreateEvent(ReArrival, Time+RandomDelay, TN, LL);
1035       DelayArray[TN] := DelayArray[TN] + RandomDelay;
1036       Inc(GCount);
1037     END
1038   END
1039 END;{ProcessReceive}
1040
1041 {*****}
1042 PROCEDURE ProcessFirstNode(FirstNode:ListPointer; VAR LL:LinkedList;
1043   VAR Q:QueueType);
1044
1045 BEGIN

```

```

1046 CASE FirstNode^.Event OF
1047   Arrival      : ProcessArrival(FirstNode, BusyTone, Terminal, LL, Q);
1048   ReArrival    : ProcessReArrival(FirstNode, BusyTone, Terminal, LL, Q);
1049   Receive      : ProcessReceive(FirstNode, BusyTone, Terminal, LL, Q);
1050 ELSE WRITELN('ERROR in EventList (ProcessFirstNode)!!!!');
1051 END;
1052
1053 END;{ProcessFirstNode}
1054
1055 {*****}
1056 PROCEDURE AdvanceNextEvent(VAR LL:LinkedList; VAR Q:QueueType);
1057 VAR
1058   FirstNode : ListPointer;
1059 BEGIN
1060   IF LL.head = NIL
1061   THEN
1062     BEGIN
1063       WRITELN('Error: EventList is empty (AdvanceNextEvent)');
1064     END
1065   ELSE IF LL.head^.NextNode = NIL
1066   THEN
1067     BEGIN
1068       FirstNode := LL.head;
1069       LL.head := NIL;
1070       ProcessFirstNode(FirstNode, LL, Q);
1071     END
1072   ELSE
1073     BEGIN
1074       FirstNode := LL.head;
1075       LL.head := LL.head^.NextNode;
1076       ProcessFirstNode(FirstNode, LL, Q);
1077     END;
1078   Dispose(FirstNode);
1079 END;{AdvanceNextEvent}
1080
1081 {*****}
1082 PROCEDURE InitEventList(VAR LL:LinkedList);
1083 VAR
1084   i : Integer;
1085   Time : Extended;
1086 BEGIN
1087   FOR i := 1 TO Maxterm DO
1088     BEGIN
1089       Time := Trunc(-(ln(1-DUNI))/ParrH) + 1;
1090       CreateEvent(Arrival, Time, i, LL);
1091     END
1092   END;{InitEventList}
1093
1094 {*****}
1095 FUNCTION CountList(VAR LL:LinkedList):Integer;
1096 VAR
1097   i : Integer;
1098   Current : ListPointer;
1099 BEGIN
1100   i := 0;

```

```

1101 Current := LL.head;
1102 WHILE Current <> NIL DO
1103 BEGIN
1104     Current := Current^.NextNode;
1105     i := i + 1;
1106 END;
1107 CountList := i;
1108 END;{CountList}
1109
1110 {*****}
1111 PROCEDURE RemoveList(VAR LL:LinkedList);
1112 VAR
1113     Current : ListPointer;
1114 BEGIN
1115     Current := LL.head;
1116     WHILE Current <> NIL DO
1117     BEGIN
1118         Current := Current^.NextNode;
1119         Dispose(LL.head);
1120         LL.head := Current;
1121     END;
1122 END;{RemoveList}
1123
1124 {*****}
1125 PROCEDURE OpenResultFile(VAR OutFile:Text);
1126 VAR
1127     GS,
1128     NL,
1129     FileName : String;
1130 BEGIN
1131     Str(GroupSize, GS);
1132     Str(NumberOfLines, NL);
1133     FileName := 'V1SLT2P7.DAT';
1134     Assign(OutFile, FileName);
1135
1136     Rewrite(OutFile);
1137 {
1138     WriteLn(OutFile, 'RESULTS FROM V1SLOT.PAS:');
1139     WriteLn(OutFile, 'Size of group : ', GroupSize);
1140     WriteLn(OutFile, 'Number of groups: ', NumberOfLines);
1141     WriteLn(OutFile);
1142     WriteLn(OutFile, 'Ptr : ', Ptr);
1143     WriteLn(OutFile, 'Alpha : ', Alpha);
1144     WriteLn(OutFile, 'SNR : ', SNR);
1145     WriteLn(OutFile);
1146     WriteLn(OutFile, 'Warm up time : ', WarmTime);
1147     WriteLn(OutFile, 'Simulation time: ', EndTime-WarmTime);
1148     WriteLn(OutFile);
1149     WriteLn(OutFile);
1150 } WriteLn(OutFile, ' G :17,
1151         'S :16,
1152         'D :16);
1153 END;{OpenResultFile}
1154
1155 {*****}
1156 { WarnBeep emits a half second warning tone and then waits 4.5 seconds }
1157 {
1158 { Depends on: none
1159 { Calls : none
1160 { Modifies : pfn
1161 {*****}
1162 Procedure WarnBeep;
1163 VAR
1164     i : Integer;
1165 BEGIN
1166     FOR i := 1 TO 5 DO
1167     BEGIN
1168         Sound(500+i*20);
1169         Delay(300);
1170     END;
1171     NoSound;
1172     Delay(1000);
1173 End;{WarnBeep}
1174
1175 {*****}
1176 { MAIN LINE
1177 {*****}
1178
1179 Begin
1180     ClrScr;
1181     WriteLn('Output Testprograma:');
1182
1183     {Initialize variables}
1184     Rreceive := 30;
1185     Rterm := 5;
1186     SNR := 20;
1187
1188     Nt := 32;
1189     NumberOfLines := 8;
1190     GroupSize := 4;
1191
1192     OpenResultFile(ResultFile);
1193     InitCodes(CCat, GroupSize, NumberOfLines);
1194     InitPathGain(S, GroupSize, NumberOfLines, Rreceive, Rterm, LossExp);
1195
1196     FOR i := 0 TO NumSim DO { Loop over terminal activity levels }
1197     BEGIN
1198         {Set terminal activity}
1199         PArrH := Power(10, -(LRangeMin + i*(LRangeMax-LRangeMin)/NumSim));
1200         rInit(3647523, 65321); {Initialize FSU Ultra random generator}
1201         InitTerm(Terminal); {Reset terminal states }
1202         InitBusyTone(BusyTone); {Set BusyTone to OFF }
1203         CreateList(EventList);
1204         CreateQueue(ReceiveQueue);
1205         InitEventList(EventList); {Initialize EventList }
1206         InitDelayArray(DelayArray); {Reset all terminal delays to zero }
1207
1208         {Show configuration to user}
1209         ClrScr;
1210         WriteLn('Receiver Radius: ', Rreceive:7:2);

```

```

1211      WriteLn('Terminal Radius: ', Rterm:7:2);
1212      WriteLn('Number of users: ', Nt:4);
1213      WriteLn('Size of group : ', GroupSize:4);
1214      WriteLn('Number of codes: ', NumberOfLines:4);
1215      WriteLn('Code length : ', codelength:4);
1216      WriteLn('Packet length : ', Packetlength:4);
1217      WriteLn;
1218      WriteLn('Terminal activity (ParrH) : ', ParrH:8:5);
1219
1220      {Reset event counters}
1221      ArrivalCount := 0;
1222      SuccessCount := 0;
1223      DelayCount := 0;
1224      GCount := 0;
1225
1226      {Warm up the network}
1227      Clock := 0;
1228      OldClock := 0;
1229      While Clock < WarmTime DO
1230      BEGIN
1231          AdvanceNextEvent(EventList, ReceiveQueue);
1232          IF (Clock - OldClock) > Interval
1233          THEN
1234              BEGIN
1235                  GoToXY(1,12);
1236                  WriteLn('Arrivals :', ArrivalCount:8);
1237                  WriteLn('Success :', SuccessCount:8);
1238                  WriteLn;
1239                  WriteLn('Delay :', DelayCount);
1240                  WriteLn('Clock :', Clock);
1241                  ShowState;
1242                  OldClock := Clock;
1243              END;
1244          GoToXY(40,4);
1245          WriteLn('SIMULATING !!!!');
1246          GoToXY(40,6);
1247          Write('STATUS: ', i, '/', NumSim);
1248          END; {Warm up loop}
1249
1250      {Reset event counters}
1251      ArrivalCount := 0;
1252      SuccessCount := 0;
1253      GCount := 0;
1254      DelayCount := 0;
1255
1256      {Main simulation Loop}
1257      While Clock < EndTime DO
1258      BEGIN
1259          AdvanceNextEvent(EventList, ReceiveQueue);
1260          IF (Clock - OldClock) > Interval
1261          THEN
1262              BEGIN
1263                  GoToXY(1,12);
1264                  WriteLn('Arrivals :', ArrivalCount:8);
1265                  WriteLn('Success :', SuccessCount:8);
1266
1267                  WriteLn;
1268                  WriteLn('Delay :', DelayCount);
1269                  WriteLn('Clock :', Clock);
1270                  ShowState;
1271                  OldClock := Clock;
1272              END;
1273          GoToXY(40,4);
1274          WriteLn('SIMULATING !!!!');
1275          GoToXY(40,6);
1276          Write('STATUS: ', i, '/', NumSim);
1277          END; {Simulation Loop}
1278
1279      {Calculate values for Physical Traffic, Throughput and Delay}
1280      Throughput := SuccessCount/(EndTime-WarmTime);
1281      D := DelayCount/SuccessCount;
1282      G := (ArrivalCount+GCount)/(EndTime-WarmTime);
1283
1284      {Write simulation results in files for processing-purposes later on}
1285      WriteLn(ResultFile, G :16,
1286              Throughput :16,
1287              D :16);
1288
1289      RemoveList(EventList);
1290
1291      END; {Terminal activity level loop}
1292
1293      Close(ResultFile);
1294
1295      {WarnBeep;}
1296      GoToXY(30,16);
1297      Write('I AM READY MISTER LE !!!!!');
1298      GoToXY(30,17);
1299      Write('Press ENTER to return to program. ');
1300      {ReadLn;}
1301      END.

```



```

1  {*****}
2  {* SimUnslotted is a program designed with the goal to simulate the *}
3  {* 'Unslotted Hybrid CDMA/ISMA protocol'. This program calculates the *}
4  {* throughput and delay of a network of terminals by simulation. *}
5  {* *}
6  {* Most of the constants, variables, functions and procedures used here *}
7  {* originates from Huub van Roosmalen. *}
8  {* *}
9  {* The terminals are clustered around distributed receivers at a fixed *}
10 {* distance. The distributed receivers are clustered around the base *}
11 {* station at a fixed distance. The packets from the terminals are *}
12 {* received by the distributed receivers and sent to the base station by *}
13 {* wire or lines. These lines can be seen as primitive concentrators *}
14 {* *}
15 {* Huy Linh Anh Le, August 10 1995 *}
16 {*****}
17 Program SimUnslotted(Input, Output);
18
19 Uses Dos, Crt, Ultrax;
20
21 Const MaxTerm = 32; { Maximum number of terminals allowed }
22 PacketLength = 64; { Number of bits per packet }
23
24 Pt = 1; { Transmitter power in Watts }
25 LossExp = 2; { Attenuation parameter for the ... }
26 { ...far field model }
27 c = 299792458; { Speed of light in m/s }
28 f = 1700000000; { Transmitter center frequency }
29 lambda = c/f; { Signal wavelength }
30
31 twoPi = 2 * Pi; { Two times Pi }
32 Pi2 = Pi * Pi; { Pi squared }
33
34 Alpha = 0.2; { The number of Rearrivals per }
35 { ...PacketDuration }
36 bitrate = 256*1024; { Bits transmitted per sec. }
37 PacketDuration = 64/bitrate; { Time needed to send a Packet }
38 Fraction = 0.2; { Inhibit delay fraction }
39
40 maxcodelength = 127; { Maximum code size }
41 codelength = 31; { Length of code }
42 Tb = 1/bitrate; { Bit duration }
43 Tc = 1/(codelength*bitrate); { Chip duration }
44 Tm = 100E-9; { Maximum delay spread }
45 MaxL = 8; { Maximum number of paths }
46 L = Trunc(Tm/Tc)+1; { Number of paths }
47
48 WarmTime = 21000; { Warm up time in Tpd }
49 Interval = 500; { Time between screen updates }
50 EndTime = 125000; { Simulation time in Tpd }
51
52 NumSim = 20; { Number of activity levels simulated }
53 LRangeMin = 4; { Terminal activity from 10^-LRangeMin }
54 LRangeMax = 0; { ...to 10^-LRangeMax }
55

```

```

56 Type TermState = (IDLE, BLOCKED, TRANSMIT, INTERFERE);
57 OnOff = (ON, OFF);
58 Matrix = Array[1..MaxTerm, 1..MaxTerm] Of Extended;
59 PosArr = Array[1..MaxTerm] Of Extended;
60 RArr = Array[1..MaxTerm, 1..MaxL] Of Extended;
61 StringArr = Array[1..MaxTerm] Of String;
62 TermArr = Array[1..MaxTerm] Of TermState;
63
64 {-----}
65 p2CRec = ^CRec;
66
67 CRec = Record
68     cc :Array[-maxcodelength..maxcodelength] Of ShortInt;
69     End;
70
71
72 CArr = Array [1..MaxTerm, 1..MaxTerm] Of p2CRec;
73
74 {-----}
75 QueueEntry= RECORD
76     TerminalNo :Integer;
77     Time :Extended;
78     END;
79
80
81 QueueType = RECORD
82     Count,
83     Front,
84     Rear: 0..Maxterm;
85     Entry: ARRAY [1..Maxterm] OF QueueEntry;
86     END;
87
88 {-----}
89 EventType = (Arrival, ReArrival, BusyToneOn, BusyToneOff, Receive);
90
91 ListPointer = ^ListNode;
92 ListNode = RECORD
93     Event :EventType;
94     Time :Extended;
95     TerminalNo :Integer;
96     NextNode :ListPointer
97     END;
98
99 LinkedList = RECORD
100     Head :ListPointer
101     END;
102
103 {-----}
104 Var Count,
105     Front,
106     Rear : Integer; { Counters for QueueType }
107
108 CCat : CArr; { Catalog of code cross correlations }
109
110 PArrH, { Packet Arrival Probability Slotted }

```

```

111   PArrL,           { Packet Arrival Probability Unslotted }
112   Rreceive,        { Receiver radius around base station }
113   Rterm,           { Terminal radius around receivers }
114   SNR              :Extended; { Signal to white Noise Ratio at receiver }
115
116   Nt,              { Number of terminals }
117   GroupSize,        { Number of terminals per group }
118   NumberOfLines :Integer; { Number of groups in the system. This is }
119   { ... also the number of codes used. }
120
121   DelayArray       :PosArr; { Delay administration for each terminal }
122   DelayCount       :Extended; { Total delay time in Tpd }
123
124   SuccessCount,    { Number of successful packets }
125   ArrivalCount,    { Number of accepted packets }
126   GCount          :LongInt; { Number of collisions ocured }
127
128   S                :Matrix; { Power loss ratio matrix }
129
130   OldClock,        { Clock maintained for screen updates }
131   Clock            :Extended; { Clock maintained by main program }
132
133   EventList        :LinkedList; { Nodes of events linked together }
134   ReceiveQueue     :QueueType; { Queue for received packets }
135
136   BusyTone         :OnOff; { Indicates wheather the channel is free }
137   { or not. }
138   Terminal         :TermArr; { Terminal state information }
139
140   i                :Integer; { Loopvariable for terminal activity levels }
141
142   ResultFile       :Text; { Result file with original values }
143
144   Throughput,      { Throughput: fraction of the time
145   { ...carrying a successful transmission }
146   D,               { Mean Delay normalized in slots }
147   G                :Extended; { Mean Delay: time between the arrival of
148   { ...the first bit till the receive of the
149   { ...last bit in sec. }
150 {*****}
151 { Power calculates a raised to the power of b }
152 {
153 { Depends on: none
154 { Calls : none
155 { Modifies : none
156 {
157 { Pre : a >= 0 And b >=0
158 { Post: Power = a^b
159 {*****}
160 Function Power(a,b:Extended):Extended;
161 Begin
162   If a<>0
163   Then
164     power := Exp(b*Ln(a))
165   Else
166     If b<>0
167     Then
168       power := 0 { power(0,b) = 0 }
169     Else
170       power := 1; { power(0,0) = 1 }
171 End;{Power}
172
173 {*****}
174 { ShowState shows the terminal states on screen }
175 {
176 { Depends on: config, numberOfLines, terminal }
177 { Calls : none
178 { Modifies : none
179 {*****}
180 Procedure ShowState;
181 Var
182   i : Integer;
183   c : Char;
184 Begin
185   GotoXY(12, 22);
186   Write(' 1 2 3 ');
187   GotoXY(12, 23);
188   Write('1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1 3 2 4 5 6 7 8 9 0 1 2');
189   For i := 1 To Nt Do
190     Begin
191       GotoXY(10+2*i,24);
192       Case Terminal[i] Of
193         IDLE : Write('i');
194         BLOCKED : Write('b');
195         TRANSMIT : Write('t');
196         INTERFERE : Write('f');
197       End;{Case}
198     End;{For i := 1 TO Nt}
199 End;{ShowState}
200
201 {*****}
202 { R calculates a partial cross correlation }
203 {
204 { Depends on: config, CCat, Tc
205 { Calls : none
206 { Modifies : none
207 {*****}
208 Function R(i,j:Integer; tau:Extended):Extended;
209 Var
210   l, { Chip number belonging to tau }
211   Cl, { Discrete correlation function for chip l }
212   Cl1 :Integer; { Discrete correlation function for chip l+1 }
213 Begin
214   l := Trunc(tau/Tc);
215   Cl1 := CCat[i,j].cc[l+1-codelength];
216   If l >= 1
217   Then
218     Begin
219       Cl := CCat[i,j].cc[l-codelength];
220       R := Cl*Tc+(Cl1-Cl)*(tau-l*Tc);

```

```

221 End
222 Else
223   R := tau*Cl1;
224 End;{R}
225
226 {*****}
227 { Rhat calculates a partial cross correlation }
228 { }
229 { Depends on: config, CCat, Tc }
230 { Calls : none }
231 { Modifies : none }
232 {*****}
233 Function Rhat(i,j:Integer; tau:Extended):Extended;
234 Var
235   l, { Chip number belonging to tau }
236   Cl, { Discrete correlation function for chip l }
237   Cl1 :Integer; { Discrete correlation function for chip l+1 }
238 Begin
239   l := Trunc(tau/Tc);
240   Cl := CCat[i,j].cc[l];
241   If l < codelength-1
242   Then
243     Begin
244       Cl1 := CCat[i,j].cc[l+1];
245       Rhat := Cl*Tc+(Cl1-Cl)*(tau-l*Tc);
246     End
247   Else
248     Rhat := ((l+1)*Tc-tau)*Cl;
249 End;{Rhat}
250
251 {*****}
252 { InitCCat sets up a catalog of discrete aperiodic cross correlation }
253 { functions for a set of (Gold) codes passed by gc. The catalog consists of }
254 { a two dimensional array of pointers. These pointers refer to }
255 { one-dimensional arrays containing the cross correlation information. This }
256 { information is in essence the number of similarities minus the numbers }
257 { of dissimilarities between two codes for a given discrete phase shift }
258 { }
259 { Depends on: config, p2CRec }
260 { Calls : none }
261 { Modifies : none }
262 {*****}
263 Procedure InitCCat(Var CCat:CArr; gc:StringArr);
264 Var
265   i, { Loop variable }
266   j, { Loop variable }
267   k, { Loop variable }
268   l, { Discrete phase shift }
269   t :Integer; { Correlation result for shift l }
270 Begin
271   Write('Calculating ', Nt, ' by ', Nt, ' correlations');
272   For i := 1 To Nt Do { Double loop over all codes }
273     For j := 1 To Nt Do
274       Begin
275         GotoXY(70,1); { Show which codes are being done }

```

```

276   Write(i:4,j:4);
277   CCat[i,j] := New(p2CRec); { Allocate space for new array }
278
279   { Calculate cross-correlation }
280   For l := -codelength To codelength Do
281     Begin
282       t := 0; { Reset correlation value }
283       If l < 0 { Different formulas for pos and }
284       Then { neg phase shifts }
285         For k := 0 To codelength-1+l Do
286           If gc[i][1+((k-l) Mod codelength)] = gc[j][k+1]
287           Then
288             Inc(t)
289           Else
290             Dec(t)
291         Else
292           For k := 0 To codelength-l-1 Do
293             If gc[i][1+(k Mod codelength)] =
294             gc[j][1+((k+l) Mod codelength)]
295             Then
296               Inc(t)
297             Else
298               Dec(t);
299           CCat[i,j].cc[l] := t;
300         End;
301       End;
302     End;{InitCCat}
303
304 {*****}
305 { DisposeCCat frees memory space taken up by the correlation catalog }
306 { }
307 { Depends on: config }
308 { Calls : none }
309 { Modifies : none }
310 {*****}
311 Procedure DisposeCCat(CCat:CArr);
312 Var
313   i, { Loop variable }
314   j :Integer; { Loop variable }
315 Begin
316   For i := 1 To Nt Do { Loop over all catalog entries }
317     For j := 1 To Nt Do
318       Dispose(CC[i,j]); { Free memory of element i,j }
319     End;{DisposeCCat}
320
321 {*****}
322 { InitTerm initializes all terminals as free }
323 { }
324 { Depends on: config }
325 { Calls : none }
326 { Modifies : pfn }
327 {*****}
328 Procedure InitTerm(Var Terminal:TermArr);
329 Var

```

```

331 i :Integer; { Loop variable }
332 Begin
333   For i:= 1 To Nt Do
334     terminal[i] := IDLE;
335 End;{InitTerm}
336
337 {*****}
338 PROCEDURE InitDelayArray(VAR DelayArray:PosArr);
339 VAR
340   i : Integer; { Loop variable }
341 BEGIN
342   FOR i := 1 TO MaxTerm DO
343     DelayArray[i] := 0;
344 END;{InitDelayArray}
345
346 {*****}
347 PROCEDURE InitBusyTone(VAR BusyTone:OnOff);
348 BEGIN
349   BusyTone := Off;
350 END;{InitBusyTone}
351
352 {*****}
353 { Initcode reads Gold codes from a file and has the cross correlation }
354 { catalog setup }
355 { }
356 { Depends on: config }
357 { Calls : initCCat }
358 { Modifies : pfn }
359 {*****}
360 Procedure InitCodes(VAR CCat:CArr; groupSize, numCodes:Integer);
361 Var
362   i, { Loop variable }
363   j :Integer; { Loop variable }
364   f :text; { File containing the Gold codes }
365   codeCount, { Length of transmission code }
366   cfile :String; { Codefile name }
367   gc :StringArr; { Gold codes as read from file }
368 Begin
369   ClrScr;
370   Str(codelength, codeCount); { Assign appropriate codefile name }
371   cfile := 'codes\' + codeCount + 'chip.dat';
372   Assign(f, cfile); { Open code file }
373   Reset(f);
374   For i := 1 To Nt Do { Read the codes from file }
375     ReadLn(f,gc[i]);
376     Close(f);
377     If numCodes <> Nt { Reassign codes according to reuse scheme }
378       Then { e.g. 12345678 -> 11223344 for 2 term/grp }
379         For i := numCodes DownTo 1 Do
380           For j := groupSize DownTo 1 Do
381             gc[(i-1)*groupSize+j] := gc[i];
382             initCCat(CCat, gc);
383             { Initialize catalog of cross correlations }
384 End;{InitCodes}
385

```

```

386 {*****}
387 { InitPathLoss sets up a table of transmitter to receiver path gains from }
388 { all transmitters to all receivers. All gains are between 0 and 1. When }
389 { groups are larger than one terminal duplicate entries will exist in the }
390 { table because there are less receivers than terminals then. }
391 { }
392 { Depends on: config, lambda }
393 { Calls : Power }
394 { Modifies : none }
395 {*****}
396 Procedure InitPathGain(VAR S:Matrix; groupSize, numberOfLines :Integer;
397   Rreceive, Rterm, lossexp:Extended );
398 Var
399   gamma, { Angle of receiver relative to x-axis }
400   theta :Extended; { Angle of transmitter relative to x-axis }
401   xt, yt, { Transmitter coordinates }
402   xr, yr :PosArr; { Receiver coordinates }
403   i, { Loop variable }
404   j, { Loop variable }
405   k :Integer; { Index variable }
406 Begin
407   For i := 1 To numberOfLines Do { Loop over all receivers }
408     For j := 1 To groupSize Do { Loop over all group terminals }
409       Begin
410         k := (i-1) * groupSize + j; {Determine index current term. }
411         gamma := 2*Pi * (i-1)/numberOfLines; {Calc. receiver angle }
412         xr[k] := Rreceive * Cos(gamma); {Calc. receiver x-coord. }
413         yr[k] := Rreceive * Sin(gamma); {Calc. receiver y-coord. }
414         theta := 2*Pi*(j-1)/groupSize+gamma+Pi; {Calc. transmitter angle }
415         xt[k] := xr[k] + Rterm * Cos(theta); {Calc. transmitter x-coord.}
416         yt[k] := yr[k] + Rterm * Sin(theta); {Calc. transmitter y-coord.}
417       End;
418       For i := 1 To Nt Do {Double loop over all terminals}
419         For j := 1 To Nt Do
420           Begin
421             { Calculate path gains according to far field model }
422             S[i,j] := Power(lambda/(4*Pi*Sqr(Sqr(xt[i]-xr[j])
423               +Sqr(yt[i]-yr[j]))),lossexp);
424           End
425 End;{InitPathGain}
426
427 {*****}
428 { Rayleigh randomly draws a value according to the rayleigh distribution }
429 { }
430 { Depends on: none }
431 { Calls : duni }
432 { Modifies : pfn }
433 {*****}
434 Function Rayleigh(ave:Extended):Extended;
435 Var
436   r,
437   u,
438   v,
439   x,
440   y :Extended;

```

```

441 Begin
442   Repeat
443     u := 2*duni-1;
444     v := 2*duni-1;
445     r := sqr(u)+sqr(v);
446   Until (r<1.0) and (r>0.0);
447   r := Sqrt(-2*Ln(r)/r);
448   x := u*r;
449   y := v*r;
450   Rayleigh:=Sqrt(Sqr(x)+Sqr(y))*ave*Sqrt(2/Pi);
451 End;
452
453 {*****}
454 { MarcumQ calculates the Marcum Q-function }
455 { }
456 { Depends on: none }
457 { Calls : none }
458 { Modifies : pfn }
459 {*****}
460 Function MarcumQ(a,b:Extended):Extended;
461 Var
462   aa,
463   ab,
464   ac,
465   bc,
466   bb,
467   ba,
468   dd,
469   d1 :Extended;
470   n :Integer;
471 Begin
472   If a = 0
473   Then
474     If b < 30
475     Then
476       MarcumQ := Exp(-Sqr(b)/2)
477     Else
478       MarcumQ := 0
479   Else
480   Begin
481     If a < b
482     Then
483       Begin
484         ab := 1;
485         dd := a/b;
486       End
487     Else
488       Begin
489         ab := 0;
490         dd := b/a;
491       End;
492     aa := 0 ;
493     bb := 0.5;
494     ba := 0 ;
495     d1 := dd;

```

```

496   n := 0;
497   Repeat
498     n := n + 1;
499     ac := d1+2*n*ab/(a*b)+aa;
500     bc := 1+2*n*bb/(a*b)+ba;
501     d1 := dd*d1;
502     aa := ab;
503     ab := ac;
504     ba := bb;
505     bb := bc;
506   Until bc > 1e12;
507   If a < b
508   Then
509     MarcumQ := (ac/(2*bc))*Exp(-Sqr(a-b)/2)
510   Else
511     MarcumQ := 1-(ac/(2*bc))*Exp(-Sqr(a-b)/2);
512   End;
513 End;
514
515 {*****}
516 { NegExpI0 approximates the modified Bessel function of the first kind }
517 { using the recurrence relation I (z) = I (z) + 2n/z I (z) }
518 { 0,n-1 0,n+1 0,n }
519 { }
520 { and then multiplies the result with Exp(-(a^2+b^2)/2). }
521 { }
522 { Depends on: none }
523 { Calls : none }
524 { Modifies : pfn }
525 {*****}
526 Function NegExpI0(a,b:Extended):Extended;
527 Var
528   n :Integer; { Loop variable }
529   x,
530   Ja, { I0,n-1 }
531   Jb, { I0,n }
532   Jc, { I0,n+1 }
533   tmp,
534   t :Extended; { Intermediate result }
535 Begin
536   tmp := (Sqr(a)+Sqr(b))/2;
537   x := a*b;
538   If x = 0
539   Then
540     If tmp < 150
541     Then
542       NegExpI0 := Exp(-tmp)/2
543     Else
544       negExpI0 := 0
545   Else
546   Begin
547     Jc := 0;
548     Jb := 1;
549     t := 0;
550     For n := 20 DownTo 1 Do

```

```

551      Begin
552          Ja := (2*n/x)*Jb+Jc;
553          Jc := Jb;
554          Jb := Ja;
555          t := t + Ja;
556      End;
557      NegExpI0 := Exp(x-(Sqr(a)+Sqr(b))/2)*Ja/(2*t-Ja);
558  End;
559 End;
560
561 {*****}
562 { CalcMu1 calculates the Mu1 parameter for the bit error formula in }
563 { Rayleigh fading environments }
564 { }
565 { Depends on: L }
566 { Calls : R, Rhat }
567 { Modifies : none }
568 {*****}
569 Function CalcMu1(u, h:Integer; tau:RArr; mu2:Extended):Extended;
570 Var
571     tmp :Extended; { Intermediate result }
572     i :Integer; { Loop variable }
573 Begin
574     tmp := 0;
575     For i := 1 To L Do
576         If i <> h
577         Then
578             tmp := tmp + R(u,u,tau[u,i])*Rhat(u,u,tau[u,i])*s[u,u];
579
580     CalcMu1 := 4*2*Pt*tmp + mu2
581 End;
582
583 {*****}
584 { CalcMu2 calculates the Mu2 parameter for the bit error formula in }
585 { Rayleigh fading environments }
586 { }
587 { Depends on: config, L, Pt, terminal }
588 { Calls : R, Rhat }
589 { Modifies : none }
590 {*****}
591 Function CalcMu2(u, h:Integer; tau:RArr; sigma_n2:Extended):Extended;
592 Var
593     tmp :Extended; { Intermediate result }
594     i, { Loop variable }
595     j :Integer; { Loop variable }
596 Begin
597     tmp := 0;
598     For i := 1 To Nt Do
599         If (terminal[i] = INTERFERE) And (i <> u)
600         Then
601             For j := 1 To L Do
602                 tmp := tmp + (Sqr(R(u,i,tau[i,j]))
603                     + Sqr(Rhat(u,i,tau[i,j])))*s[i,u];
604
605     For j := 1 To L Do

```

```

606     If j <> h
607     Then
608         tmp := tmp + (Sqr(R(u,u,tau[u,j]))
609             + Sqr(Rhat(u,u,tau[u,j])))*s[u,u];
610
611     CalcMu2 := 2*(2*Pt*tmp+sigma_n2);
612 End;
613
614 {*****}
615 { CalcMu12 calculates the Mu12 parameter for the bit error formula in }
616 { Rayleigh fading environments }
617 { }
618 { Depends on: config, L, Pt, terminal }
619 { Calls : R, Rhat }
620 { Modifies : none }
621 {*****}
622 Function CalcMu12(u,h:Integer;tau:RArr):Extended;
623 Var
624     tmp :Extended; { Intermediate result }
625     i, { Loop variable }
626     j :Integer; { Loop variable }
627 Begin
628     tmp := 0;
629     For i := 1 To Nt Do
630         If (terminal[i] = INTERFERE) And (i<>u)
631         Then
632             For j := 1 To L Do
633                 tmp := tmp + R(u,i,tau[i,j])*Rhat(u,i,tau[i,j])*s[i,u];
634
635     For j := 1 To L Do
636         If j <> h
637         Then
638             tmp := tmp + (Sqr(Rhat(u,u,tau[u,j]))
639                 + R(u,u,tau[u,j])*Rhat(u,u,tau[u,j]))*s[u,u];
640
641     CalcMu12 := 2*2*Pt*tmp;
642 End;
643
644 {*****}
645 { CalculatePbe computes the bit error probability for a terminal, given }
646 { which other terminals transmit }
647 { }
648 { Depends on: config, L, Pt, S, Tb, terminal }
649 { Calls : duni, I0, MarcumQ, Rayleigh }
650 { Modifies : none }
651 {*****}
652 Function CalculatePbe(u:Integer):Extended;
653 Var
654     mu1,
655     mu2,
656     mu12,
657     a,
658     b,
659     bmax, { Maximum path gain }
660     tmp,

```

```

661 m,
662 mq,
663 mb,
664 sigma_n2:Extended;      { Noise power }
665 beta,                    { Rayleigh distributed path gains }
666 tau :RArr;               { Uniformly [0,Tb] distributed path delays }
667 h,                        { Number of path with largest gain }
668 i,                        { Loop variable }
669 j :Integer;               { Loop variable }
670 Begin
671 { Assign random delays and path gains to transmitting terminals }
672 For i := 1 To Nt Do
673   If terminal[i] = INTERFERE
674   Then
675     For j := 1 To L Do
676       Begin
677         tau[i,j] := Tb*duni;
678         beta[i,j] := Rayleigh(Sqrt(S[i,u]));
679       End
680     Else
681       For j := 1 To L Do
682         Begin
683           tau[i,j] := 0;
684           beta[i,j] := 0;
685         End;
686
687 { Determine what the largest gain is }
688 bmax := 0;
689 h := 1;
690 For i := 1 To L Do
691   If beta[u,i]>bmax
692   Then
693     Begin
694       bmax := beta[u,i];
695       h := i;
696     End;
697
698 { Determine noise power relative to largest incoming signal }
699 sigma_n2 := power(10, -snr/10)*Sqr(bmax)*Pt*Tb*Tb;
700
701 { Calculate bit error probability parameters }
702 mu2 := CalcMu2(u, h, tau, sigma_n2);
703 mu1 := CalcMu1(u, h, tau, mu2);
704 mu12 := CalcMu12(u, h, tau);
705 m := Sqrt(2*Pt)*bmax*Tb;
706 a := m*Abs(1/Sqrt(mu1)-1/Sqrt(mu2))/Sqrt(2);
707 b := m*(1/Sqrt(mu1)+1/Sqrt(mu2))/Sqrt(2);
708 mq := MarcumQ(a,b);
709 mb := NegExp10(a,b);
710
711 { Calculate the bit error probability }
712 tmp := ( (1 + mu12/Sqrt(mu1*mu2))*mb )/2;
713 CalculatePbe := mq-tmp
714 End;
715

```

```

716 {*****}
717 {      Q U E U E S      }
718 {*****}
719
720 {*****}
721 { CreateQueue }
722 { Pre : None }
723 { Post: The queue Q has been created and is initialized to be empty. }
724 {*****}
725 PROCEDURE CreateQueue(VAR Q:QueueType);
726 BEGIN
727   WITH Q DO
728     BEGIN
729       Count := 0;
730       Front := 1;
731       Rear := 0
732     END
733   END;{CreateQueue}
734
735 {*****}
736 PROCEDURE Append (TerminalNo:Integer; Time:Extended; VAR Q:QueueType);
737 VAR
738   X : QueueEntry;
739 BEGIN
740   WITH Q DO
741     WITH Q DO
742       IF Count = MaxTerm THEN
743         WRITELN ('Error: Attempt to append an entry to a full queue')
744       ELSE
745         BEGIN
746           X.TerminalNo := TerminalNo;
747           X.Time := Time;
748           Count := Count + 1;
749           Rear := (Rear MOD MaxTerm) + 1;
750           Entry [Rear] := X;
751         END
752       END;{Append}
753
754 {*****}
755 PROCEDURE Remove (VAR X: QueueEntry; VAR Q:QueueType);
756 BEGIN
757   WITH Q DO
758     IF Count = 0 THEN
759       WRITELN ('Error: Attempt to remove an entry from an empty queue')
760     ELSE
761       BEGIN
762         Count := Count - 1;
763         X := Entry [Front];
764         Front := (Front MOD MaxTerm) + 1
765       END
766     END;{Remove}
767
768 {*****}
769 FUNCTION QueueSize (VAR Q: QueueType): INTEGER;
770 BEGIN

```



```

771 QueueSize := Q.Count
772 END;{QueueSize}
773
774 {*****}
775 FUNCTION QueueEmpty (VAR Q: QueueType): BOOLEAN;
776 BEGIN
777     QueueEmpty := (Q.Count = 0)
778 END;{QueueEmpty}
779
780 {*****}
781 FUNCTION QueueFull (VAR Q: QueueType): BOOLEAN;
782 BEGIN
783     QueueFull := (Q.Count = MaxTerm)
784 END;{QueueFull}
785
786 {*****}
787 PROCEDURE ShowQueue (VAR Q : QueueType);
788 VAR
789     x, y : INTEGER;
790     X1 : QueueEntry;
791 BEGIN
792     {WRITELN('De inhoud van de queue is:');}
793     x := QueueSize (Q);
794     WITH Q DO
795         y := front;
796         WHILE x > 0 DO
797             BEGIN
798                 X1 := Q.Entry[y];
799                 WRITELN('TN: ',X1.TerminalNo,'Time: ',X1.Time);
800                 x := x-1;
801                 y := (y MOD MaxTerm) + 1
802             END;
803             WRITELN;
804 END;{ShowQueue}
805
806
807 {*****}
808 { LINKED - LIST }
809 {*****}
810
811
812 {*****}
813 {* CreateList initializes linked list LL to be empty *}
814 {* *}
815 {* Pre: None *}
816 {* Post: The list LL has been created and is initialized to be empty *}
817 {*****}
818 PROCEDURE CreateList(VAR LL: LinkedList);
819 BEGIN
820     LL.head := NIL
821 END;{CreateList}
822
823 {*****}
824 PROCEDURE InsertEvent(NewNode:ListPointer; T:Extended; VAR LL:LinkedList);
825 VAR

```

```

826 Current,
827 Trailing : ListPointer;
828 BEGIN
829     Current := LL.head;
830     Trailing := NIL;
831     WHILE (Current^.Time <= T) AND (Current^.NextNode <> NIL) DO
832         BEGIN
833             Trailing := Current;
834             Current := Current^.NextNode;
835         END;
836     IF (Current^.time <= T) AND (Current^.NextNode = NIL)
837     THEN
838         BEGIN
839             Current^.NextNode := NewNode;
840             NewNode^.NextNode := NIL;
841         END
842     ELSE IF (Current^.Time > T) AND (Trailing = NIL)
843     THEN
844         BEGIN
845             NewNode^.NextNode := LL.head;
846             LL.head := NewNode;
847         END
848     ELSE
849         BEGIN
850             NewNode^.NextNode := Current;
851             Trailing^.NextNode := NewNode;
852         END;
853 END;{InsertEvent}
854
855 {*****}
856 PROCEDURE CreateEvent(E:EventType; T:Extended; TN:Integer; VAR LL:LinkedList);
857
858 VAR
859     NewNode : ListPointer;
860 BEGIN
861     New(NewNode);
862     NewNode^.Event := E;
863     NewNode^.Time := T;
864     NewNode^.TerminalNo := TN;
865     IF LL.head = NIL
866     THEN
867         BEGIN
868             NewNode^.NextNode := NIL;
869             LL.head := NewNode;
870         END
871     ELSE
872         BEGIN
873             InsertEvent(NewNode, T, LL);
874         END;
875 END;{CreateEvent}
876
877 {*****}
878 PROCEDURE ShowList(VAR LL:LinkedList);
879 VAR
880     Current : ListPointer;

```



```

881 i      : Integer;
882 Event  : String;
883 BEGIN
884     i      := 1;
885     Current := LL.head;
886     ClrScr;
887     WriteLn('The list consists of: (Press ENTER for the nodes to show up)');
888     WriteLn;
889     IF Current = NIL THEN
890         BEGIN
891             WRITELN('EventList is empty! (WriteList)');
892         END
893     ELSE
894         WHILE (Current <> NIL) DO
895             BEGIN
896                 CASE Current^.Event OF
897                     Arrival   : Event := 'Arrival';
898                     ReArrival : Event := 'ReArrival';
899                     BusyToneOn : Event := 'BusyToneOn';
900                     BusyToneOff : Event := 'BusyToneOff';
901                     Receive    : Event := 'Receive';
902                 ELSE WRITELN('ERROR in EventList (ShowList)!!!!');
903             END;
904             WRITELN('Node:', i, '      Event:', Event, '12,'      Time:',
905                 Current^.Time, 4, '      Terminal:', Current^.TerminalNo);
906             i := i+1;
907             Current := Current^.NextNode;
908             ReadLn;
909         END;
910         WriteLn('End of list');
911         ReadLn;
912     END; (ShowList)
913
914 {*****}
915 PROCEDURE ProcessArrival(FirstNode:ListPointer; VAR BusyTone:OnOff;
916     VAR Terminal:TermArr; VAR LL:LinkedList;
917     VAR Q:QueueType);
918 VAR
919     RandomDelay,
920     Time,
921     TInterArr : Extended;
922     TN        : Integer;
923 BEGIN
924     Clock      := FirstNode^.Time;
925     TN         := FirstNode^.TerminalNo;
926     Time       := FirstNode^.Time;
927     TInterArr := -(ln(1-DUNI))/ParrH;
928     CreateEvent(Arrival, Time + TInterArr, TN, LL);
929     IF Terminal[TN] = IDLE
930     THEN
931         BEGIN
932             Terminal[TN] := BLOCKED;
933             Inc(ArrivalCount);
934             IF BusyTone = ON
935             THEN

```

```

936         BEGIN
937             RandomDelay := -(ln(1-DUNI))/Alpha;
938             CreateEvent(ReArrival, Time + RandomDelay, TN, LL);
939             DelayArray[TN] := DelayArray[TN] + RandomDelay;
940             Inc(GCount);
941         END
942     ELSE
943         BEGIN
944             Terminal[TN] := TRANSMIT;
945             Append(TN, Time, Q);
946             CreateEvent(BusyToneOn, Time+Fraction, 0, LL);
947             CreateEvent(Receive, Time+1+Fraction, TN, LL);
948             DelayArray[TN] := DelayArray[TN] + 1 + Fraction;
949         END
950     END
951 END; (ProcessArrival)
952
953 {*****}
954 PROCEDURE ProcessReArrival(FirstNode:ListPointer; VAR BusyTone:OnOff;
955     VAR Terminal:TermArr; VAR LL:LinkedList;
956     VAR Q:QueueType);
957 VAR
958     RandomDelay,
959     Time,
960     TInterArr : Extended;
961     TN        : Integer;
962 BEGIN
963     Clock      := FirstNode^.Time;
964     TN         := FirstNode^.TerminalNo;
965     Time       := FirstNode^.Time;
966     IF BusyTone = ON
967     THEN
968         BEGIN
969             RandomDelay := -(ln(1-DUNI))/Alpha;
970             CreateEvent (ReArrival, Time + RandomDelay, TN, LL);
971             DelayArray[TN] := DelayArray[TN] + RandomDelay;
972             Inc(GCount);
973         END
974     ELSE
975         BEGIN
976             Terminal[TN] := TRANSMIT;
977             Append(TN, Time, Q);
978             CreateEvent(BusyToneOn, Time+Fraction, 0, LL);
979             CreateEvent(Receive, Time+1+Fraction, TN, LL);
980             DelayArray[TN] := DelayArray[TN] + 1 + Fraction;
981         END
982     END; (ProcessReArrival)
983
984 {*****}
985 PROCEDURE ProcessBusyToneOn(FirstNode:ListPointer; VAR BusyTone:OnOff;
986     VAR LL:LinkedList);
987 BEGIN
988     Clock := FirstNode^.Time;
989     BusyTone := ON;
990 END; (ProcessBusyToneOn)

```

```

991
992 {*****}
993 PROCEDURE ProcessBusyToneOff (FirstNode:ListPointer; VAR BusyTone:OnOff);
994 BEGIN
995     Clock := FirstNode^.Time;
996     BusyTone := OFF;
997 END;{ProcessBusyToneOff}
998
999 {*****}
1000 PROCEDURE ProcessReceive(FirstNode:ListPointer; VAR BusyTone:OnOff;
1001     VAR Terminal:TermArr; VAR LL:LinkedList;
1002     VAR Q:QueueType);
1003 VAR
1004     X2      : QueueEntry;
1005
1006     TN,
1007     TN2     : Integer;
1008
1009     RandomDelay,
1010     Time,
1011     Time2,
1012     Pbe,
1013     Psp     : Extended;
1014
1015 BEGIN
1016     Clock := FirstNode^.Time;
1017     Time := FirstNode^.Time;
1018     IF QueueEmpty(Q) = TRUE
1019     THEN
1020     BEGIN
1021         TN := FirstNode^.TerminalNo;
1022         RandomDelay := -(ln(1-DUNI))/Alpha;
1023         Terminal[TN] := BLOCKED;
1024         CreateEvent(ReArrival, Time+RandomDelay, TN, LL);
1025         DelayArray[TN] := DelayArray[TN] + RandomDelay;
1026         Inc(GCount);
1027     END
1028     ELSE
1029     BEGIN
1030         TN := FirstNode^.TerminalNo;
1031         Terminal[TN] := INTERFERE;
1032         WHILE NOT QueueEmpty(Q) DO
1033         BEGIN
1034             Remove(X2, Q);
1035             TN2 := X2.TerminalNo;
1036             Time2 := X2.Time;
1037             Terminal[TN2] := INTERFERE;
1038         END;
1039         Pbe := CalculatePbe(TN);
1040         Psp := Power(1-Pbe, PacketLength);
1041         CreateEvent(BusyToneOff, Time2+1+Fraction, 0, LL);
1042
1043         IF DUNI < Psp
1044         THEN
1045             BEGIN
1046                 Inc(SuccessCount);
1047                 Terminal[TN] := IDLE;
1048                 DelayCount := DelayCount + DelayArray[TN];
1049                 DelayArray[TN] := 0;
1050             END
1051             ELSE
1052             BEGIN
1053                 Terminal[TN] := BLOCKED;
1054                 RandomDelay := -(ln(1-DUNI))/Alpha;
1055                 CreateEvent(ReArrival, Time+RandomDelay, TN, LL);
1056                 DelayArray[TN] := DelayArray[TN] + RandomDelay;
1057                 Inc(GCount);
1058             END
1059             END
1060         END;{ProcessReceive}
1061
1062 {*****}
1063 PROCEDURE ProcessFirstNode(FirstNode:ListPointer; VAR LL:LinkedList;
1064     VAR Q:QueueType);
1065
1066 BEGIN
1067     CASE FirstNode^.Event OF
1068     Arrival      : ProcessArrival(FirstNode, BusyTone, Terminal, LL, Q);
1069     ReArrival    : ProcessReArrival(FirstNode, BusyTone, Terminal, LL, Q);
1070     BusyToneOn   : ProcessBusyToneOn(FirstNode, BusyTone, LL);
1071     BusyToneOff  : ProcessBusyToneOff(FirstNode, BusyTone);
1072     Receive      : ProcessReceive(FirstNode, BusyTone, Terminal, LL, Q);
1073     ELSE WRITELN('ERROR in EventList (ProcessFirstNode)!!!!');
1074     END;
1075
1076 END;{ProcessFirstNode}
1077
1078 {*****}
1079 PROCEDURE AdvanceNextEvent(VAR LL:LinkedList; VAR Q:QueueType);
1080 VAR
1081     FirstNode : ListPointer;
1082 BEGIN
1083     IF LL.head = NIL
1084     THEN
1085     BEGIN
1086         WRITELN('Error: EventList is empty (AdvanceNextEvent)');
1087     END
1088     ELSE IF LL.head^.NextNode = NIL
1089     THEN
1090     BEGIN
1091         FirstNode := LL.head;
1092         LL.head := NIL;
1093         ProcessFirstNode(FirstNode, LL, Q);
1094     END
1095     ELSE
1096     BEGIN
1097         FirstNode := LL.head;
1098         LL.head := LL.head^.NextNode;
1099         ProcessFirstNode(FirstNode, LL, Q);
1100     END;

```

```

1101 Dispose(FirstNode);
1102 END;{AdvanceNextEvent}
1103
1104 {*****}
1105 PROCEDURE InitEventList(VAR LL:LinkedList);
1106 VAR
1107     i : Integer;
1108     Time : Extended;
1109 BEGIN
1110     FOR i := 1 TO Maxterm DO
1111     BEGIN
1112         Time := -(ln(1-DUNI))/ParrH;
1113         CreateEvent(Arrival, Time, i, LL);
1114     END
1115 END;{InitEventList}
1116
1117 {*****}
1118 FUNCTION CountList(VAR LL:LinkedList):Integer;
1119 VAR
1120     i : Integer;
1121     Current : ListPointer;
1122 BEGIN
1123     i := 0;
1124     Current := LL.head;
1125     WHILE Current <> NIL DO
1126     BEGIN
1127         Current := Current^.NextNode;
1128         i := i + 1;
1129     END;
1130     CountList := i;
1131 END;{CountList}
1132
1133 {*****}
1134 PROCEDURE RemoveList(VAR LL:LinkedList);
1135 VAR
1136     Current : ListPointer;
1137 BEGIN
1138     Current := LL.head;
1139     WHILE Current <> NIL DO
1140     BEGIN
1141         Current := Current^.NextNode;
1142         Dispose(LL.head);
1143         LL.head := Current;
1144     END;
1145 END;{RemoveList}
1146
1147 {*****}
1148 PROCEDURE OpenResultFile(VAR OutFile:Text);
1149 VAR
1150     GS,
1151     NL,
1152     FileName : String;
1153 BEGIN
1154     Str(GroupSize, GS);
1155     Str(NumberOfLines, NL);
1156     FileName := 'V2H8W3.DAT';
1157     Assign(OutFile, FileName);
1158
1159     Rewrite(OutFile);
1160     WriteLn(OutFile, 'Size of group : ', GroupSize);
1161     WriteLn(OutFile, 'Number of groups: ', NumberOfLines);
1162     WriteLn(OutFile);
1163     WriteLn(OutFile, 'Inhibit delay fraction: ', Fraction);
1164     WriteLn(OutFile, 'Alpha : ', Alpha);
1165     WriteLn(OutFile, 'SNR : ', SNR);
1166     WriteLn(OutFile);
1167 }
1168     WriteLn(OutFile, ' G :17,
1169             'S :16,
1170             'D :16);
1171 END;{OpenResultFile}
1172
1173 {*****}
1174 { WarnBeep emits a half second warning tone and then waits 4.5 seconds }
1175 {
1176 { Depends on: none
1177 { Calls : none
1178 { Modifies : pfn
1179 {*****}
1180 Procedure WarnBeep;
1181 VAR
1182     i : Integer;
1183 BEGIN
1184     FOR i := 1 TO 5 DO
1185     BEGIN
1186         Sound(500+i*20);
1187         Delay(300);
1188     END;
1189     NoSound;
1190     Delay(1000);
1191 END;{WarnBeep}
1192
1193 {*****}
1194 { MAIN LINE }
1195 {*****}
1196
1197 Begin
1198     ClrScr;
1199     WriteLn('Output Testprograma:');
1200
1201     {Initialize variables}
1202     Rreceive := 30;
1203     Rterm := 5;
1204     SNR := 6;
1205
1206     Nt := 32;
1207     NumberOfLines := 4;
1208     GroupSize := 8;
1209
1210     OpenResultFile(ResultFile);

```

```

1211 InitCodes(CCat, GroupSize, NumberOfLines);
1212 InitPathGain(S, GroupSize, NumberOfLines, Rreceive, Rterm, LossExp);
1213
1214 FOR i := 0 TO NumSim DO      { Loop over terminal activity levels }
1215 BEGIN
1216   {Set terminal activity}
1217   PArrH := Power(10, -(LRangeMin + i*(LRangeMax-LRangeMin)/NumSim));
1218   rInit(3647523, 65321);    {Initialize FSU Ultra random generator}
1219   InitTerm(Terminal);      {Reset terminal states }
1220   InitBusyTone(BusyTone);  {Set BusyTone to OFF }
1221   CreateList(EventList);
1222   CreateQueue(ReceiveQueue);
1223   InitEventList(EventList); {Initialize EventList }
1224   InitDelayArray(DelayArray); {Reset all terminal delays to zero }
1225
1226   {Show configuration to user}
1227   ClrScr;
1228   WriteLn('Receiver Radius: ', Rreceive:7:2);
1229   WriteLn('Terminal Radius: ', Rterm:7:2);
1230   WriteLn('Number of users: ', Nt:4);
1231   WriteLn('Size of group : ', GroupSize:4);
1232   WriteLn('Number of codes: ', NumberOfLines:4);
1233   WriteLn('Code length : ', codelength:4);
1234   WriteLn('Packet length : ', Packetlength:4);
1235   WriteLn;
1236   WriteLn('Terminal activity (PArrH) : ', PArrH:8:5);
1237
1238   {Reset event counters}
1239   ArrivalCount := 0;
1240   SuccessCount := 0;
1241   DelayCount := 0;
1242   GCount := 0;
1243
1244   {Warm up the network}
1245   Clock := 0;
1246   OldClock := 0;
1247   While Clock < WarmTime DO
1248   BEGIN
1249     AdvanceNextEvent(EventList, ReceiveQueue);
1250     IF (Clock - OldClock) > Interval
1251     THEN
1252     BEGIN
1253       GoToXY(1,12);
1254       WriteLn('Arrivals : ', ArrivalCount:8);
1255       WriteLn('Success : ', SuccessCount:8);
1256       WriteLn;
1257       WriteLn('Delay : ', DelayCount);
1258       WriteLn('Clock : ', Clock);
1259       ShowState;
1260       OldClock := Clock;
1261     END;
1262     GoToXY(40,4);
1263     WriteLn('SIMULATING !!!!');
1264     GoToXY(40,6);
1265     Write('STATUS: ', i, '/', NumSim);
1266
1267     END;{Warm up loop}
1268
1269     {Reset event counters}
1270     ArrivalCount := 0;
1271     SuccessCount := 0;
1272     GCount := 0;
1273     DelayCount := 0;
1274
1275     {Main simulation Loop}
1276     While Clock < EndTime DO
1277     BEGIN
1278       AdvanceNextEvent(EventList, ReceiveQueue);
1279       IF (Clock - OldClock) > Interval
1280       THEN
1281       BEGIN
1282         GoToXY(1,12);
1283         WriteLn('Arrivals : ', ArrivalCount:8);
1284         WriteLn('Success : ', SuccessCount:8);
1285         WriteLn;
1286         WriteLn('Delay : ', DelayCount);
1287         WriteLn('Clock : ', Clock);
1288         ShowState;
1289         OldClock := Clock;
1290       END;
1291       GoToXY(40,4);
1292       WriteLn('SIMULATING !!!!');
1293       GoToXY(40,6);
1294       Write('STATUS: ', i, '/', NumSim);
1295     END;{Simulation Loop}
1296
1297     {Calculate values for Physical Traffic, Throughput and Delay}
1298     Throughput := SuccessCount/(EndTime-WarmTime);
1299     D := DelayCount/SuccessCount;
1300     G := (ArrivalCount+GCount)/(EndTime-WarmTime);
1301
1302     {Write simulation results in files for processing-purposes later on}
1303     WriteLn(ResultFile, G :16,
1304             Throughput :16,
1305             D :16);
1306
1307     RemoveList(EventList);
1308
1309   END;{Terminal activity level loop}
1310
1311   Close(ResultFile);
1312
1313   {WarnBeep;}
1314   GoToXY(30,16);
1315   Write('I AM READY MISTER LE !!!!!');
1316   GoToXY(30,17);
1317   Write('Press ENTER to return to program. ');
1318   {ReadLn;}
1319 END.

```

```

1 {*****}
2 {* V2_CALC1.pas is a program designed with the goal to calculate the *}
3 {* performance of the 'Unslotted Hybrid CDMA/ISMA protocol'. This program *}
4 {* calculates the throughput and delay of a network of terminals by *}
5 {* a mathematical formula. *}
6 {* *}
7 {* Most of the constants, variables, functions and procedures used here *}
8 {* originates from Huub van Roosmalen. *}
9 {* *}
10 {* The terminals are clustered around distributed receivers at a fixed *}
11 {* distance. The distributed receivers are clustered around the base *}
12 {* station at a fixed distance. The packets from the terminals are *}
13 {* received by the distributed receivers and sent to the base station by *}
14 {* wire or lines. These lines can be seen as primitive concentrators. *}
15 {* *}
16 {* Most time parameters are normalized to the PacketDuration. To have a *}
17 {* time parameter in seconds one needs to multiply by the PacketDuration *}
18 {* *}
19 {* Huy Linh Anh Le, July 16 1994 *}
20 {*****}
21
22 Program CalcUnslotted(Input, Output);
23
24 Uses Dos, Crt, Ultrax;
25
26 Const MonteCarlo = 1000;
27 MaxTerm = 32; { Maximum number of terminals allowed }
28 XMax = 4; { Maximum number of interference pkcts }
29 PacketLength = 64; { Number of bits per packet }
30
31 Pt = 1; { Transmitter power in Watts }
32 LossExp = 2; { Attenuation parameter for the ... }
33 { ...far field model }
34 c = 299792458; { Speed of light in m/s }
35 f = 1700000000; { Transmitter center frequency }
36 lambda = c/f; { Signal wavelength }
37
38 twoPi = 2 * Pi; { Two times Pi }
39 Pi2 = Pi * Pi; { Pi squared }
40
41 Alpha = 0.1; { x retransmission per PacketDuration }
42 Fraction = 0.01; { Inhibit delay fraction (d) }
43
44 bitrate = 256*1024; { Bits transmitted per sec. }
45 PacketDuration = 64/bitrate; { Time needed to send a Packet }
46 maxcodelength = 127; { Maximum code size }
47 Codelength = 31; { Length of code }
48 Tb = 1/bitrate; { Bit duration }
49 Tc = 1/(codelength*bitrate); { Chip duration }
50 Tm = 100E-9; { Maximum delay spread }
51 MaxL = 8; { Maximum number of paths }
52 L = Trunc(Tm/Tc)+1; { Number of paths }
53
54 NumSim = 20;
55 LRangeMin = 3;

```

```

56 LRangeMax = -2;
57
58 Type TermState = (IDLE, BLOCKED, TRANSMIT, INTERFERE);
59 Matrix = Array[1..MaxTerm, 1..MaxTerm] Of Extended;
60 PosArr = Array[1..MaxTerm] Of Extended;
61 RArr = Array[1..MaxTerm, 1..MaxL] Of Extended;
62 StringArr = Array[1..MaxTerm] Of String;
63 TermArr = Array[1..MaxTerm] Of TermState;
64 ConfigArr = Array[1..XMax] Of Integer;
65 PpsArr = Array[1..XMax] Of Extended;
66 ExtArr = Array[0..XMax] Of Extended;
67 SArr = Array[0..NumSim] Of Extended;
68 {-----}
69 p2CRec = ^CRec;
70
71 CRec = Record
72 cc : Array[-maxcodelength..maxcodelength] Of ShortInt;
73 End;
74
75
76 CArr = Array [1..MaxTerm, 1..MaxTerm] Of p2CRec;
77
78
79 Var CCat : CArr; { Catalog of code cross correlations }
80
81 Rreceive, { Receiver radius around base station }
82 Rterm, { Terminal radius around receivers }
83 SNR :Extended; { Signal to white Noise Ratio at receiver }
84
85 Nt, { Number of terminals }
86 GroupSize, { Number of terminals per group }
87 NumberOfLines :Integer; { Number of groups in the system. This is }
88 { ... also the number of codes used. }
89
90 S :Matrix; { Power loss ratio matrix }
91
92 Terminal :TermArr; { Terminal state information }
93
94 X,
95 i,
96 j :Integer; { Loopvariable for terminal activity levels }
97
98 CountConfig :LongInt;
99 StartConfig,
100 MaxConfig :ConfigArr;
101 PpsAveXArr :PpsArr;
102 SArray :SArr;
103
104 ResultFile :Text;
105
106 PpsAveX,
107 PpsXZero,
108 PConfl,
109 PConflTemp,
110 PNoConfl,

```

```

111 ExpGd,
112 Poisson,
113 TCycle,
114 Pidle,
115 Psuccess,
116 Throughput,
117 D,
118 G      : Extended;
119
120 Faculteit : ExtArr;
121
122 {*****}
123 { Power calculates a raised to the power of b }
124 { }
125 { Depends on: none }
126 { Calls : none }
127 { Modifies : none }
128 { }
129 { Pre : a >= 0 And b >= 0 }
130 { Post: Power = a^b }
131 {*****}
132 Function Power(a,b:Extended):Extended;
133 Begin
134   If a<>0
135   Then
136     power := Exp(b*Ln(a))
137   Else
138     If b<>0
139     Then
140       power := 0 { power(0,b) = 0 }
141     Else
142       power := 1; { power(0,0) = 1 }
143 End;{Power}
144
145 {*****}
146 { InitFact }
147 {*****}
148 Procedure InitFact (Var fact:ExtArr);
149 Var
150   i : Integer;
151 Begin
152   fact[0] := 1;
153   For i := 1 TO XMax DO
154     fact[i] := i*fact[i-1];
155 End;{InitFact}
156
157 {*****}
158 { ShowState shows the terminal states on screen }
159 { }
160 { Depends on: config, numberOfLines, terminal }
161 { Calls : none }
162 { Modifies : none }
163 {*****}
164 Procedure ShowState;
165 Var

```

```

166   i : Integer;
167   c : Char;
168 Begin
169   GotoXY(12, 22);
170   Write('          1          2          3 ');
171   GotoXY(12, 23);
172   Write('1 2 3 4 5 6 7 8 9 0 1 2 3 4 5 6 7 8 9 0 1 3 2 4 5 6 7 8 9 0 1 2');
173   For i := 1 To Nt Do
174   Begin
175     GotoXY(10+2*i,24);
176     Case Terminal[i] Of
177       IDLE      : Write('i');
178       BLOCKED   : Write('b');
179       TRANSMIT  : Write('t');
180       INTERFERE : Write('f');
181     End;{Case}
182   End;{For i := 1 TO Nt}
183 End;{ShowState}
184
185 {*****}
186 { R calculates a partial cross correlation }
187 { }
188 { Depends on: config, CCat, Tc }
189 { Calls : none }
190 { Modifies : none }
191 {*****}
192 Function R(i,j:Integer; tau:Extended):Extended;
193 Var
194   l, { Chip number belonging to tau }
195   Cl, { Discrete correlation function for chip l }
196   Cl1 :Integer; { Discrete correlation function for chip l+1 }
197 Begin
198   l := Trunc(tau/Tc);
199   Cl1 := CCat[i,j]^cc[l+1-codelength];
200   If l >= 1
201   Then
202   Begin
203     Cl := CCat[i,j]^cc[l-codelength];
204     R := Cl*Tc+(Cl1-Cl)*(tau-l*Tc);
205   End
206   Else
207     R := tau*Cl1;
208 End;{R}
209
210 {*****}
211 { Rhat calculates a partial cross correlation }
212 { }
213 { Depends on: config, CCat, Tc }
214 { Calls : none }
215 { Modifies : none }
216 {*****}
217 Function Rhat(i,j:Integer; tau:Extended):Extended;
218 Var
219   l, { Chip number belonging to tau }
220   Cl, { Discrete correlation function for chip l }

```

```

221 Cl1      :Integer;      { Discrete correlation function for chip l+1 }
222 Begin
223   l      := Trunc(tau/Tc);
224   Cl     := CCat[i,j]^cc[l];
225   If l < codelength-1
226   Then
227   Begin
228     Cl1 := CCat[i,j]^cc[l+1];
229     Rhat := Cl*Tc+(Cl1-Cl)*(tau-l*Tc);
230   End
231   Else
232     Rhat := ((l+1)*Tc-tau)*Cl;
233 End;(Rhat)
234
235 {*****}
236 { InitCCat sets up a catalog of discrete aperiodic cross correlation }
237 { functions for a set of (Gold) codes passed by gc. The catalog consists of }
238 { a two dimensional array of pointers. These pointers refer to }
239 { one-dimensional arrays containing the cross correlation information. This }
240 { information is in essence the number of similarities minus the numbers }
241 { of dissimilarities between two codes for a given discrete phase shift }
242 { }
243 { Depends on: config, p2CRec }
244 { Calls      : none }
245 { Modifies   : none }
246 {*****}
247 Procedure InitCCat(Var CCat:CArr; gc:StringArr);
248 Var
249   i,      { Loop variable }
250   j,      { Loop variable }
251   k,      { Loop variable }
252   l,      { Discrete phase shift }
253   t      :Integer; { Correlation result for shift l }
254 Begin
255   Write('Calculating ', Nt, ' by ', Nt, ' correlations');
256   For i := 1 To Nt Do { Double loop over all codes }
257     For j := 1 To Nt Do
258       Begin
259         GotoXY(70,1); { Show which codes are being done }
260         Write(i:4,j:4);
261         CCat[i,j] := New(p2CRec); { Allocate space for new array }
262
263         { Calculate cross-correlation }
264         For l := -codelength To codelength Do
265           Begin
266             t := 0; { Reset correlation value }
267             If l < 0 { Different formulas for pos and }
268             Then { neg phase shifts }
269               For k := 0 To codelength-1+l Do
270                 If gc[i][1+((k-l) Mod codelength)] = gc[j][k+1]
271                 Then
272                   Inc(t)
273                 Else
274                   Dec(t)
275             Else

```

```

276       For k := 0 To codelength-l-1 Do
277         If gc[i][1+((k Mod codelength))] =
278           gc[j][1+((k+l) Mod codelength)]
279         Then
280           Inc(t)
281         Else
282           Dec(t);
283
284       CCat[i,j]^cc[l] := t;
285     End;
286   End;
287 End;(InitCCat)
288
289 {*****}
290 { DisposeCCat frees memory space taken up by the correlation catalog }
291 { }
292 { Depends on: config }
293 { Calls      : none }
294 { Modifies   : none }
295 {*****}
296 Procedure DisposeCCat(CCat:CArr);
297 Var
298   i,      { Loop variable }
299   j      :Integer; { Loop variable }
300 Begin
301   For i := 1 To Nt Do { Loop over all catalog entries }
302     For j := 1 To Nt Do
303       Dispose(CCat[i,j]); { Free memory of element i,j }
304 End;(DisposeCCat)
305
306 {*****}
307 { InitTerm initializes all terminals as free }
308 { }
309 { Depends on: config }
310 { Calls      : none }
311 { Modifies   : pfn }
312 {*****}
313 Procedure InitTerm(Var Terminal:TermArr);
314 Var
315   i :Integer; { Loop variable }
316 Begin
317   For i:= 1 To Nt Do
318     terminal[i] := IDLE;
319 End;(InitTerm)
320
321 {*****}
322 { Initcode reads Gold codes from a file and has the cross correlation }
323 { catalog setup }
324 { }
325 { Depends on: config }
326 { Calls      : initCCat }
327 { Modifies   : pfn }
328 {*****}
329 Procedure InitCodes(Var CCat:CArr; groupSize, numCodes:Integer);
330 Var

```



```

331 i,                { Loop variable }
332 j                :Integer;        { Loop variable }
333 f                :text;           { File containing the Gold codes }
334 codeCount,       { Length of transmission code }
335 cfile            :String;         { Codefile name }
336 gc               :StringArr;      { Gold codes as read from file }
337 Begin
338   ClrScr;
339   Str(codeLength, codeCount); { Assign appropriate codefile name }
340   cfile := 'codes\' + codeCount + 'chip.dat';
341   Assign(f, cfile);           { Open code file }
342   Reset(f);
343   For i := 1 To Nt Do         { Read the codes from file }
344     ReadLn(f, gc[i]);
345   Close(f);
346   If numCodes <> Nt           { Reassign codes according to reuse scheme }
347     Then                      { e.g. 12345678 -> 11223344 for 2 term/grp }
348       For i := numCodes DownTo 1 Do
349         For j := groupSize DownTo 1 Do
350           gc[(i-1)*groupSize+j] := gc[i];
351           initCCat(CCcat, gc);
352           { Initialize catalog of cross correlations }
353 End; {InitCodes}
354
355 {*****}
356 { InitPathLoss sets up a table of transmitter to receiver path gains from }
357 { all transmitters to all receivers. All gains are between 0 and 1. When }
358 { groups are larger than one terminal duplicate entries will exist in the }
359 { table because there are less receivers than terminals then. }
360 { }
361 { Depends on: config, lambda }
362 { Calls      : Power }
363 { Modifies   : none }
364 {*****}
365 Procedure InitPathGain(Var S:Matrix; groupSize, numberOfLines :Integer;
366                       Rreceive, Rterm, lossexp:Extended );
367 Var
368   gamma,            { Angle of receiver relative to x-axis }
369   theta             :Extended; { Angle of transmitter relative to x-axis }
370   xt, yt,           { Transmitter coordinates }
371   xr, yr            :PosArr;   { Receiver coordinates }
372   i,                { Loop variable }
373   j,                { Loop variable }
374   k                 :Integer;   { Index variable }
375 Begin
376   For i := 1 To numberOfLines Do { Loop over all receivers }
377     For j := 1 To groupSize Do { Loop over all group terminals }
378       Begin
379         k := (i-1) * groupSize + j; {Determine index current term. }
380         gamma := 2*Pi * (i-1)/numberOfLines; {Calc. receiver angle }
381         xr[k] := Rreceive * Cos(gamma); {Calc. receiver x-coord. }
382         yr[k] := Rreceive * Sin(gamma); {Calc. receiver y-coord. }
383         theta := 2*Pi*(j-1)/groupSize+gamma+Pi; {Calc. transmitter angle }
384         xt[k] := xr[k] + Rterm * Cos(theta); {Calc. transmitter x-coord.}
385         yt[k] := yr[k] + Rterm * Sin(theta); {Calc. transmitter y-coord.}
386       End;
387     End;
388   For i := 1 To Nt Do
389     For j := 1 To Nt Do
390       Begin
391         { Calculate path gains according to far field model }
392         S[i,j] := Power(lambda/(4*Pi*Sqrt(Sqr(xt[i]-xr[j])
393           +Sqr(yt[i]-yr[j]))), lossexp);
394       End;
395     End; {InitPathGain}
396   {*****}
397   {* InitStartConfig }
398   {*****}
399   PROCEDURE InitStartConfig (VAR Start:ConfigArr; X:Integer);
400   VAR
401     i : Integer;
402   BEGIN
403     FOR i := 1 TO X DO
404       Start[i] := i+1;
405   END; {InitStartConfig}
406   {*****}
407   {* InitMaxConfig }
408   {*****}
409   PROCEDURE InitMaxConfig (VAR Max:ConfigArr; X:Integer);
410   VAR
411     i : Integer;
412   BEGIN
413     FOR i := X DOWNTO 1 DO
414       Max[i] := Nt-(X-i);
415   END; {InitMaxConfig}
416   {*****}
417   {* WriteConfig }
418   {*****}
419   PROCEDURE WriteConfig (Var Config:ConfigArr; X:Integer);
420   VAR
421     i : Integer;
422   BEGIN
423     Write('Configuration = ');
424     For i := 1 TO X DO
425       Write(Config[i], ' ');
426     WriteLn;
427   END; {WriteConfig}
428   {*****}
429   {* Rayleigh randomly draws a value according to the rayleigh distribution }
430   {*****}
431   { Depends on: none }
432   { Calls      : duni }
433   { Modifies   : pfn }
434   {*****}
435   Function Rayleigh(ave:Extended):Extended;
436   Var
437     r,

```



```

441 u,
442 v,
443 x,
444 y      :Extended;
445 Begin
446   Repeat
447     u := 2*duni-1;
448     v := 2*duni-1;
449     r := sqrt(u)+sqrt(v);
450   Until (r<1.0) and (r>0.0);
451   r := Sqrt(-2*Ln(r)/r);
452   x := u*r;
453   y := v*r;
454   Rayleigh:=Sqrt(Sqr(x)+Sqr(y))*ave*Sqrt(2/Pi);
455 End;
456
457 {*****}
458 { MarcumQ calculates the Marcum Q-function }
459 { }
460 { Depends on: none }
461 { Calls      : none }
462 { Modifies   : pfn }
463 {*****}
464 Function MarcumQ(a,b:Extended):Extended;
465 Var
466   aa,
467   ab,
468   ac,
469   bc,
470   bb,
471   ba,
472   dd,
473   d1      :Extended;
474   n        :Integer;
475 Begin
476   If a = 0
477   Then
478     If b < 30
479     Then
480       MarcumQ := Exp(-Sqr(b)/2)
481     Else
482       MarcumQ := 0
483   Else
484     Begin
485       If a < b
486       Then
487         Begin
488           ab := 1;
489           dd := a/b;
490         End
491       Else
492         Begin
493           ab := 0;
494           dd := b/a;
495         End;

```

```

496   aa := 0 ;
497   bb := 0.5;
498   ba := 0 ;
499   d1 := dd;
500   n := 0;
501   Repeat
502     n := n + 1;
503     ac := d1+2*n*ab/(a*b)+aa;
504     bc := 1+2*n*bb/(a*b)+ba;
505     d1 := dd*d1;
506     aa := ab;
507     ab := ac;
508     ba := bb;
509     bb := bc;
510   Until bc > 1e12;
511   If a < b
512   Then
513     MarcumQ := (ac/(2*bc))*Exp(-Sqr(a-b)/2)
514   Else
515     MarcumQ := 1-(ac/(2*bc))*Exp(-Sqr(a-b)/2);
516   End;
517 End;
518
519 {*****}
520 { NegExpI0 approximates the modified Bessel function of the first kind }
521 { using the recurrence relation I (z) = I (z) + 2n/z I (z) }
522 { 0,n-1 0,n+1 0,n }
523 { }
524 { and then multiplies the result with Exp(-(a^2+b^2)/2). }
525 { }
526 { Depends on: none }
527 { Calls      : none }
528 { Modifies   : pfn }
529 {*****}
530 Function NegExpI0(a,b:Extended):Extended;
531 Var
532   n      :Integer; { Loop variable }
533   x,
534   Ja,
535   Jb,
536   Jc,
537   tmp,
538   t      :Extended; { Intermediate result }
539 Begin
540   tmp := (Sqr(a)+Sqr(b))/2;
541   x := a*b;
542   If x = 0
543   Then
544     If tmp < 150
545     Then
546       NegExpI0 := Exp(-tmp)/2
547     Else
548       negExpI0 := 0
549   Else
550     Begin

```

```

551 Jc := 0;
552 Jb := 1;
553 t := 0;
554 For n := 20 DownTo 1 Do
555 Begin
556   Ja := (2*n/x)*Jb+Jc;
557   Jc := Jb;
558   Jb := Ja;
559   t := t + Ja;
560 End;
561 NegExp10 := Exp(x-(Sqr(a)+Sqr(b))/2)*Ja/(2*t-Ja);
562 End;
563 End;
564
565 {*****}
566 { CalcMu1 calculates the Mu1 parameter for the bit error formula in }
567 { Rayleigh fading environments }
568 { }
569 { Depends on: L }
570 { Calls : R, Rhat }
571 { Modifies : none }
572 {*****}
573 Function CalcMu1(u, h:Integer; tau:RArr; mu2:Extended):Extended;
574 Var
575   tmp :Extended; { Intermediate result }
576   i :Integer; { Loop variable }
577 Begin
578   tmp := 0;
579   For i := 1 To L Do
580     If i <> h
581     Then
582       tmp := tmp + R(u,u,tau[u,i])*Rhat(u,u,tau[u,i])*s[u,u];
583   End;
584   CalcMu1 := 4*2*Pt*tmp + mu2;
585 End;
586
587 {*****}
588 { CalcMu2 calculates the Mu2 parameter for the bit error formula in }
589 { Rayleigh fading environments }
590 { }
591 { Depends on: config, L, Pt, terminal }
592 { Calls : R, Rhat }
593 { Modifies : none }
594 {*****}
595 Function CalcMu2(u, h:Integer; tau:RArr; sigma_n2:Extended):Extended;
596 Var
597   tmp :Extended; { Intermediate result }
598   i, { Loop variable }
599   j :Integer; { Loop variable }
600 Begin
601   tmp := 0;
602   For i := 1 To Nt Do
603     If (terminal[i] = INTERFERE) And (i <> u)
604     Then
605       For j := 1 To L Do

```

```

606       tmp := tmp + (Sqr(R(u,i,tau[i,j]))
607         + Sqr(Rhat(u,i,tau[i,j])))*s[i,u];
608   End;
609   For j := 1 To L Do
610     If j <> h
611     Then
612       tmp := tmp + (Sqr(R(u,u,tau[u,j]))
613         + Sqr(Rhat(u,u,tau[u,j])))*s[u,u];
614   End;
615   CalcMu2 := 2*(2*Pt*tmp+sigma_n2);
616 End;
617
618 {*****}
619 { CalcMu12 calculates the Mu12 parameter for the bit error formula in }
620 { Rayleigh fading environments }
621 { }
622 { Depends on: config, L, Pt, terminal }
623 { Calls : R, Rhat }
624 { Modifies : none }
625 {*****}
626 Function CalcMu12(u,h:Integer;tau:RArr):Extended;
627 Var
628   tmp :Extended; { Intermediate result }
629   i, { Loop variable }
630   j :Integer; { Loop variable }
631 Begin
632   tmp := 0;
633   For i := 1 To Nt Do
634     If (terminal[i] = INTERFERE) And (i<>u)
635     Then
636       For j := 1 To L Do
637         tmp := tmp + R(u,i,tau[i,j])*Rhat(u,i,tau[i,j])*s[i,u];
638   End;
639   For j := 1 To L Do
640     If j <> h
641     Then
642       tmp := tmp + (Sqr(Rhat(u,u,tau[u,j]))
643         + R(u,u,tau[u,j])*Rhat(u,u,tau[u,j]))*S[u,u];
644   End;
645   CalcMu12 := 2*2*Pt*tmp;
646 End;
647
648 {*****}
649 { CalculatePbe computes the bit error probability for a terminal, given }
650 { which other terminals transmit }
651 { }
652 { Depends on: config, L, Pt, S, Tb, terminal }
653 { Calls : duni, IO, MarcumQ, Rayleigh }
654 { Modifies : none }
655 {*****}
656 Function CalculatePbe(u:Integer):Extended;
657 Var
658   mu1,
659   mu2,
660   mu12,

```

```

661 a,
662 b,
663 bmax,          { Maximum path gain }
664 tmp,
665 m,
666 mq,
667 mb,
668 sigma_n2:Extended; { Noise power }
669 beta,          { Rayleigh distributed path gains }
670 tau :RArr;     { Uniformly [0,Tb] distributed path delays }
671 h,            { Number of path with largest gain }
672 i,            { Loop variable }
673 j            { Loop variable }
674 Begin
675 { Assign random delays and path gains to transmitting terminals }
676 For i := 1 To Nt Do
677   If terminal[i] = INTERFERE
678   Then
679     For j := 1 To L Do
680       Begin
681         tau[i,j] := Tb*duni;
682         beta[i,j] := Rayleigh(Sqrt(S[i,u]));
683       End
684     Else
685       For j := 1 To L Do
686         Begin
687           tau[i,j] := 0;
688           beta[i,j] := 0;
689         End;
690
691 { Determine what the largest gain is }
692 bmax := 0;
693 h := 1;
694 For i := 1 To L Do
695   If beta[u,i]>bmax
696   Then
697     Begin
698       bmax := beta[u,i];
699       h := i;
700     End;
701
702 { Determine noise power relative to largest incoming signal }
703 sigma_n2 := power(10, -snr/10)*Sqr(bmax)*Pt*Tb*Tb;
704
705 { Calculate bit error probability parameters }
706 mu2 := CalcMu2(u, h, tau, sigma_n2);
707 mu1 := CalcMu1(u, h, tau, mu2);
708 mu12 := CalcMu12(u, h, tau);
709 m := Sqrt(2*Pt)*bmax*Tb;
710 a := m*Abs(1/Sqrt(mu1)-1/Sqrt(mu2))/Sqrt(2);
711 b := m*(1/Sqrt(mu1)+1/Sqrt(mu2))/Sqrt(2);
712 mq := MarcumQ(a,b);
713 mb := NegExpI0(a,b);
714
715 { Calculate the bit error probability }

```

```

716 tmp := ( (1 + mu12/Sqrt(mu1*mu2))*mb )/2;
717 CalculatePbe := mq-tmp
718 End;
719
720 {*****}
721 {* CalcPpsXZero *}
722 {*****}
723 FUNCTION CalcPpsXZero : Extended;
724 VAR
725   i : Integer;
726   PpsTot : Extended;
727 BEGIN
728   InitTerm(Terminal);
729   Terminal[1] := INTERFERE;
730   PpsTot := 0;
731   If L=1
732   Then
733     CalcPpsXZero := Power(1-CalculatePbe(1), PacketLength)
734   Else
735     Begin
736       For i := 1 To MonteCarlo Do
737         BEGIN
738           PpsTot := PpsTot + Power(1-CalculatePbe(1), PacketLength);
739         END;
740       CalcPpsXZero := PpsTot/MonteCarlo;
741     End
742 End;{CalcPpsXZero}
743
744 {*****}
745 {* CalcPpsAveX *}
746 {*****}
747 FUNCTION CalcPpsAveX (X:Integer):Extended;
748 VAR
749   i,ii,j : Integer;
750   Pbe,
751   PpsTot : Extended;
752   TempConfig : ConfigArr;
753 BEGIN
754   InitTerm(Terminal);
755   Terminal[1] := INTERFERE;
756   TempConfig := StartConfig;
757   PpsTot := 0;
758   CountConfig := 0;
759   FOR ii := 1 TO X DO
760     Terminal[TempConfig[ii]] := INTERFERE;
761   FOR ii := 1 TO MonteCarlo DO
762     Begin
763       Pbe := CalculatePbe(1);
764       PpsTot := PpsTot + Power(1-Pbe, PacketLength);
765     End;
766   CountConfig := CountConfig + 1;
767   {WriteConfig(TempConfig, X);}
768   i := X;
769   WHILE (i<>1) OR (TempConfig[1]<>MaxConfig[1]) DO
770     BEGIN

```

```

771 i := X;
772 IF TempConfig[i] = Nt
773 THEN
774 BEGIN
775     WHILE TempConfig[i] = MaxConfig[i] DO
776         i := i-1;
777         TempConfig[i] := TempConfig[i] + 1;
778         FOR j := (i+1) TO X DO
779             TempConfig[j] := TempConfig[j-1] + 1;
780         END
781     ELSE
782         TempConfig[i] := TempConfig[i] + 1;
783     InitTerm(Terminal);
784     Terminal[1] := INTERFERE;
785     FOR ii := 1 TO X DO
786         Terminal[TempConfig[ii]] := INTERFERE;
787     ShowState;
788     FOR ii := 1 TO MonteCarlo DO
789         Begin
790             Pbe := CalculatePbe(1);
791             PpsTot := PpsTot + Power(1-Pbe, PacketLength);
792         End;
793         CountConfig := CountConfig + 1;
794         {WriteConfig(TempConfig, X);}
795     END;
796     CalcPpsAveX := PpsTot/(MonteCarlo*CountConfig);
797     {WriteLn(X, ' ', CountConfig, ' ', MonteCarlo*CountConfig);}
798 END;{CalcPpsAveX}
799
800 {*****}
801 PROCEDURE OpenResultFile(VAR OutFile:Text);
802 VAR
803     GS,
804     NL,
805     FileName : String;
806 BEGIN
807     Str(GroupSize, GS);
808     Str(NumberOfLines, NL);
809     FileName := '2SDK7L2.dat';
810     Assign(OutFile, FileName);
811
812     Rewrite(OutFile);
813 {
814     WriteLn(OutFile, 'H.L.A. Le');
815     WriteLn(OutFile, 'RESULTS FROM CALCULATIONS (v.2):');
816     WriteLn(OutFile);
817     WriteLn(OutFile, 'Receiver Radius: ', Rreceive);
818     WriteLn(OutFile, 'Terminal Radius: ', Rterm);
819     WriteLn(OutFile);
820     WriteLn(OutFile, 'Number of users : ', Nt);
821     WriteLn(OutFile, 'Size of group : ', GroupSize);
822     WriteLn(OutFile, 'Number of groups: ', NumberOfLines);
823     WriteLn(OutFile);
824     WriteLn(OutFile, 'Code length : ', codelength);
825     WriteLn(OutFile, 'Packet length : ', Packetlength);
826     WriteLn(OutFile, 'Inhibit delay fraction: ', Fraction);

```

```

826     WriteLn(OutFile, 'SNR : ', SNR);
827     WriteLn(OutFile);
828     WriteLn(OutFile);}
829 END;{OpenResultFile}
830
831 {*****}
832 { WarnBeep emits a half second warning tone and then waits 4.5 seconds }
833 {
834 { Depends on: none
835 { Calls : none
836 { Modifies : pfn
837 {*****}
838 PROCEDURE WarnBeep;
839 VAR
840     i : Integer;
841 BEGIN
842     FOR i := 1 TO 5 DO
843         BEGIN
844             Sound(500+i*20);
845             Delay(300);
846         END;
847         NoSound;
848         Delay(1000);
849     End;{WarnBeep}
850
851 {*****}
852 { M A I N L I N E }
853 {*****}
854 Begin
855     ClrScr;
856     WriteLn('Output Testprograma:');WriteLn;
857
858     {Initialize Variables}
859     Rreceive := 30;
860     Rterm := 5;
861     SNR := 6;
862
863     Nt := 32;
864     NumberOfLines := 2;
865     GroupSize := 16;
866
867     { Initialize Functions and Procedures }
868     InitCodes(CCat, GroupSize, NumberOfLines);
869     InitPathGain(S, GroupSize, NumberOfLines, Rreceive, Rterm, LossExp);
870     InitFact(Faculteit);
871     WriteLn('NumberOfLines: ',NumberOfLines);
872     WriteLn;
873
874     { Calculate PpsAveX }
875     OpenResultFile(ResultFile);
876     FOR X := 1 TO XMax Do
877         BEGIN
878             InitStartConfig(StartConfig, X);
879             InitMaxConfig(MaxConfig, X);
880             PpsAveX := CalcPpsAveX(X);

```

```

881      PpsAveXArr[X] := PpsAveX;
882      WriteLn(ResultFile, PpsAveX);
883      Flush(ResultFile);
884  END;
885  WriteLn(ResultFile);
886  WriteLn(ResultFile, ' G:          ':24,
887                'S(4):          ':23,
888                'D(4):          ':23);
889
890  { Throughput and Delay Calculation }
891  PpsXZero := CalcPpsXZero;
892  For i := 0 To NumSim Do
893  Begin
894      G      := Power(10, -(LRangeMin+i*(LRangeMax-LRangeMin)/NumSim));
895      ExpGd   := Exp(-G*fraction);
896      TCycle  := (1+(2*fraction))*(ExpGd/G);
897      PNoConfl := ExpGd*PpsXZero;
898      Pidle   := 1/(G*(1+(2*fraction))+ExpGd);
899      PConflTemp := 0;
900      Write(ResultFile, G:23);
901      For j := 1 To XMax Do
902      Begin
903          Poisson := Power(G*fraction,j) * ExpGd/Faculteit[j];
904          PConflTemp := PConflTemp + Poisson*PpsAveXArr[j];
905      End;
906      PConfl := PConflTemp;
907      Psuccess := PNoConfl+PConfl;
908      Throughput := Psuccess/TCycle;
909      D := (((G*Pidle)/Throughput)-1)*(1+fraction+(1/Alpha))
910          +(G*(1-Pidle)/Throughput)*(1/Alpha)
911          +(1+fraction);
912      Write(ResultFile, Throughput:23);
913      WriteLn(ResultFile, D:23);
914  End;
915  Close(ResultFile);
916
917
918  {WarnBeep;}
919  WriteLn;
920  WriteLn;
921  Write('I AM READY MISTER LE !!!!!');
922  GoToXY(35,17);
923  Write('Press ENTER to return to program. ');
924  {ReadLn;}
925  END.

```