#### An Ultra-Low-Power ADPLL for BLE Applications

Analysis, Design and Implementation

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Analysis, Design and Implementation

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Master of Science Thesis

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#### Simplicity is the ultimate sophistication

Leonardo da Vinci

万物之始,大道至简,衍化至繁 老子

## Abstract

In recent years, wireless personal area network (WPAN) applications have triggered the needs for low-cost and low-power PLLs which also provide good performance. All-digital phased-locked loops (ADPLLs) are preferred over their analog counterparts in nanoscale CMOS technology due to their flexibility, configurability, small area and easy portability. However, fractional spurs and insufficiently low power dissipation are main problems related to conventional TDC-based structures. In this work, a sub-half mW 2.2 GHz - 3 GHz fractional-N ADPLL is presented for Bluetooth Low Energy (BLE) applications. Coarse-fine DTC based phase predictor with dynamic element matching (DEM) ability and clock gated phase error freezer are proposed to reduce the power while maintaining good phase noise and fractional spur performance. This prototype ADPLL was taped out on Sep. 11th 2014 in GlobalFoundries 40 nm Low Power (40 nm-LP) technology. Based on post-layout simulations and modelling, it is expected to consume less than 450  $\mu$ W with integrated rms jitter of 1.5 ps for the close integer channel and 800 fs for the rest of channels, leading to a potential state-of-art FoM below -240 dB. Design of the full ADPLL in terms of system level analysis, digital logic, mixed-signal and RF design is presented in the thesis.

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together two years ago, after bachelor study. In a word I owe my accomplishments to my family.

Lianbo Wu Delft, 2014

## Glossary

### List of Acronyms

	CIUIIYIIIS		
ADPLL	all-digital phase-locked loop		
CP-PLL	charge-pump phase-locked loop		
CKR	retimed reference clock		
СКУ	variable clock		
ckvd2	divide-by-2 variable clock		
CMOS	complementary metal-oxide-semiconductor		
DCO	digitally controlled oscillator		
DTC	digital-to-time converter		
FCW	frequency command word		
FoM	figure-of-merit		
отw	oscillator tuning word		
РСВ	printed circuit board		
TDC	time-to-digital converter		
МоМ	Metal-oxide-Metal		
DEM	dynamic element matching		
TSPC	true single phase clock		
SPI	serial peripheral interface		
WPAN	wireless personal area network		

- **WSN** wireless sensor network
- LMS least mean squares
- LFSR linear feedback shift register
- TX transmitter
- **INL** integral nonlinearity
- **DNL** differential linearity
- **PVS** Cadence physical verification system
- **CML** current-mode logic

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1

## Introduction

Yes, we have to divide up our time like that, between our politics and our equations. But to me our equations are far more important, for politics are only a matter of present concern. A mathematical equation stands forever.

Albert Einstein

#### 1.1. Introduction

hat frequency synthesizer to IC is what watch to human being. Frequency synthesizer is such an important as well as indispensable item that you can see it everywhere in modern IC products. From telecommunication systems to digital circuit applications, from clock and data recovery to modulation and waveform generation, frequency synthesizer takes places in almost every IC circuit, either wireless or wireline. In the mean time, remarkably growth has been seen in wireless communication ever since year 1901 when Marconi succeed in his radio experiment, and nowadays people are enjoying the reality of anytime anyplace communication thanks to emerging IC products. In this emerging IC market, many untapped opportunities exist in the realm of short range low-cost wireless networks. Applications in wireless personal area network (WPAN) and wireless sensor network (WSN) are virgin territories where the new era of wireless would be triggered. Power budget analysis of such sensor nodes reveals that the wireless transceiver dominates the overall power. This, together with the need for large volume, low-cost, highly integrated solutions warrant the need for ultra-low power RF transceivers in nanometer scale digital CMOS technologies. This thesis is an effort in that direction and explores the implementation of an ultra-low power all-digital phase-locked loop (ADPLL) for frequency synthesis in WPAN transceivers with main focus on Bluetooth Low Energy (BLE) application.

# **1.2.** Frequency Synthesizer for Telecommunication Systems and Modulation

Frequency synthesizer is intensively wanted in telecommunication applications. It provides the means by which a communication or broadcast channel can be selected. A wireless radio is such a system, which enables the flow of information from one point to another, usually through air. Here Figure 1.1 shows a simplified block diagram of a typical low-power transceiver. A low-IF receiver and direct modulation transmitter is typically the preferred architecture. In a low-IF receiver, the received radio frequency (RF) signal from the antenna is amplified by the low-noise amplifier (LNA) after down-converted to a low intermediate frequency (IF) by a quadrature mixer. Following that, low pass filter and programmable gain amplifier (PGA) is used to feed the signal into an analog-to-digital converter (ADC). The

digitized data would be processed further by digital baseband. On the transmitter side, the digital baseband converts the user data to symbols which are finally transformed to polar form with amplitude and phase components. The frequency phase data then modulates the synthesizer while the amplitude data controls the power amplifier. Conclusion could be easily drawn here that a frequency synthesizer not only plays a key role to generate local oscillator for mixer in receiver but also contributes significantly to frequency modulation in transmitter path.



Figure 1.1: Example transceiver architecture

#### **1.3.** Common Metrics for Frequency Synthesizer

As shown in Figure 1.1, the synthesizer is a very important part of the radio. It must meet very demanding specifications for their performance. These include:

**Frequency tuning range and resolution** The synthesizer must be able to cover all frequencies necessary to cover every channel of interest according to the application. Besides, the adjacent programmable frequencies of the synthesizer must be separated by no more than the channel spacing of the radio.

**Acquisition time** This is the settling time after channel switch. In another word, the time it takes to switch from one channel to another with a required accuracy. This can be crucial for some applications requiring fast switching such as frequency hopping spread spectrum. Moreover, much more power is consumed before the synthesizer settles down and a short switching time is really important

for duty-cycled low power systems.

**Freedom from spurs** The waveform must be free of spurs, or spurious frequency components other than the one desired. In every standard there is a requirement about the amplitude of the spur tones to be at least a specified decibels lower than the desired carrier.

**Purity of the output tone** Generally, in addition to the desired tone, there will be additional unwanted frequency. The spreading of the tone power in the frequency domain around the desired frequency is often called phase noise. In time domain it could be related as time jitter. Both the spurious tones and the noise skirt degrade the performance of the transceiver. In time domain the output of an ADPLL could be expressed as

$$v(t) = A\sin(\omega_c t + \psi(t)) \tag{1.1}$$

Where A represents the amplitude of the synthesized signal while  $\omega_c$  is its frequency.  $\psi$  captures the phase fluctuation and it could be expressed as

$$\psi(t) = \Delta \psi \sin(\omega_m t) + \phi(t) \tag{1.2}$$

The first item represents the periodical modulation and will appear as spurious tone at the output. The second item in the equation is the random fluctuation and it is shown as noise-skirt around the carrier frequency.

#### 1.4. PLL based Frequency synthesizer

Although other methods such as direct digital synthesis exit, phase-locked loop (PLL) based frequency synthesis is by far the most popular one and widely used in communication systems owing to its feature of generating high frequency with low phase noise. A generic PLL diagram is shown in Figure 1.2. Essentially it is a negative feedback system which contains basically three core parts: (1) a phase detector (PD), (2) a loop filter and (3) an oscillator. The oscillator is the soul as well as engine to generate a high frequency according to the input reference clock with a certain ratio. The phase of the output variable signal is compared with the input reference clock by the phase detector and the phase error is passed to the loop filter. The filtered phase error signal is processed to control the oscillator in the direction that the phase error is reduced. When the loop is finally locked, the average output frequency equals exactly to the input signal under that certain



Figure 1.2: Essential PLL diagram

ratio. The output is then tracking the input signal. The history of PLL dates back to early 1930s when British researchers developed the zero-IF receiver to alternate the famous superheterodyne one proposed by Armstrong in 1918. Problem was about the rapid drift in frequency with the output of the local oscillator. A French engineer Henri de Bellescize published the paper [1] about the idea of maintaining the phase of the oscillator in a desired way by adding a feedback correction. This is recognized as the first PLL ever published. In 1960s, boosted by large needs in consumer products such as analog TV and explosion of IC industry, PLL theory and design got more mature and numerous analog as well as digital PLL structures were explored and published ever since then. Nevertheless, all-digital PLL did not come to life until early this century. In the year of 2004, a novel all digital PLL system was published by TI [2] and it triggered a new trend of PLL design, serving as the start point of this thesis.

#### 1.4.1. Charge-pump PLL

Figure 1.3 shows a simplified block diagram of the conventional Charge-Pump PLL (CP-PLL). The oscillator part is implemented as a voltage-controlled oscillator (VCO) in an analog way. The phase frequency detector (PFD) compares the phase of the reference signal and the divided down output signal and generates an up/down pulse with a width proportional to the phase error. The output of the phase detector controls the magnitude and direction of the charge pump current which is pumped into or out of the loop filter. The passive loop-filter converts this current into a tuning voltage for the VCO while suppressing the noise from the reference signal and the phase frequency detector.

CP-PLL itself is a good architecture for WPAN application in terms of its power and phase noise performance. However, the analog intensive nature of the CP-PLL makes its implementation difficult in advanced process nodes. Voltage headroom



Figure 1.3: Simplified block diagram of charge-pump PLL

is reduce in advanced technology nodes due to the reduced voltage supply and this is a bad news for CP-PLL who basically operates in voltage domain. Besides, the passives do not scale well with the technology and the loop-filter used in CP-PLL occupies large area, increasing the overall system costs. In addition, the reduced gate-oxide thickness in the nanoscale CMOS technologies results in significant current leakage through the integration capacitor of the loop-filter. This leakage current increases the total PLL jitter, thereby degrades the performance.

#### **1.5.** Motivation

As analog frequency synthesizer design comes across problems in advanced CMOS process, methods to integrate RF front-end with baseband processor are highly desired. Compared with conventional analog design, a digital intensive approach provides other benefits such as better testability, more reconfigurability, smaller area and higher degree of integration. Besides, it is also easier for digital calibration such as DCO gain calibration or fractional spur reducing calibration.

At this proper moment, ADPLL's emergence meets this anticipation and also enables the way leading to digital assisted RF, e.g. all digital transmitter with two-point modulation. Compared with its analog counterpart, it is not only more economic but also technology scaling friendly. Its performance is already proved in standards

#### 1.5. Motivation

such from Bluetooth to GSM [3] to be able to alternate the CP-PLL. What's more, the settling procedure is generally more faster owing to the digital algorithm whom it bases on. Numerous publications and researches have been conducted to optimize the performance regarding power, phase noise and spur level. However, it is still difficult to see a DPLL design realized with sub-mW power consumption. As shown in Figure 1.4, the only design to break sub-mW barrier among all the state of the art in the past decade [2] [4] [5] [6] [7] [8] [9] is proposed by IMEC Holst Centre [5]. Nevertheless, in terms of phase noise and power there is still a large headroom to fully explore. This is also the start point as well as motivation for this thesis work, which is to reduced the worst fractional spur with optimized phase noise and power. The target specifications will be analysed in detail in Chapter 2.



Figure 1.4: FoM for recent year designs <sup>1</sup>.

<sup>1.</sup> The figure of merit (FoM) is defined as  $FoM = 10 \log[\frac{\sigma_{jitter}}{1s}^2 * (\frac{P}{1mW})][10]$ 

#### **1.6.** Research Contribution

The time table which records my master thesis design is shown in Figure 1.5. As depicted in the following chapters, a first-ever coarse-fine phase prediction based ADPLL design with sub-half mW and optimized fractional spur level is proposed. According to the post-layout simulation and verified model, a state of the art FoM is expected. This is a one-year one-student project at NXP semiconductor implemented in GlobalFoundries 40 nm-LP technology. My theoretical work mainly lies in the analysis of fractional spur and rough investigation about mismatch in inverter based DTC. Nevertheless, the main contribution of mine still comes from the implementation of a full ADPLL. I am in charge of the system level analysis, modelling of the PLL, core digital logic and SPI RTL coding as well as front-end (specification to netlist) synthesizing, circuit level implementation of DCO, divider, counter, clock gating and DTC. I got access to IMEC Holst Centre's design data [5] as a reference in late January this year in TSMC 40 nm and did three times of measurement in IMEC before March due to administration arrangement. The design is sent for tapeout on Sep. 11th and the I am preparing the PCB for measurement. I will finish the measurement at NXP also.

#### **1.7.** Outline of the Thesis

This thesis presents the design of a full ADPLL from system level to transistorlevel. It is organized as follows. Chapter 2 begins with a background introduction of ADPLL architecture. After that, fractional spur issue is analysed in terms of reason, level estimation as well as proposed solution. Based on the proposed solution, a coarse-fine phase predictor based ultra-low power ADPLL system is proposed. Chapter 3 clarifies the simulation method I used and the modelling of work of this ADPLL design I have done. The implementation of the whole ADPLL is introduced in Chapter 4 from three aspects: digital logic, mixed signal design and RF design. Chapter 5 summarizes the top level simulation result while Chapter 6 draws the final conclusion of this work. The appendix includes the important data not shown in the bulk part of the thesis, like the register table, some direct simulation results as well as some related details of derivations described in the main part.



Figure 1.5: Master thesis timeline.

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# 2

## ADPLL System: Background, Analysis and Proposed Architecture

All-digital PLL (ADPLL) structure vouched its potential for ultra-low power application in nanoscale CMOS technology in recent publications [1], as its conventional analog counterpart finds more difficulties to scale in CMOS technology. As one of the most exciting invention in RF field, counter based ADPLL not only addresses the issues such as reference spur faced by charge-pump PLL, but also benefits people with more reconfigurability, flexibility features to meet the stringent requirements set by the modern advanced wireless standards.

Based on the background introduction given previously, this chapter provides further investigation on ADPLL design from system aspect. Section 2.1 outlines the current popular Digital PLL architectures. Section 2.2 offers basic background analysis of ADPLL system from both frequency and time perspectives. Fractional spur issue is explored in depth in section 2.3. Section 2.4 clarifies the specifications of this thesis based on Bluetooth Low Energy standard. Section 2.5 proposes a novel ADPLL architecture to realize low noise and small spur at low power.

#### **2.1.** Current Popular Architectures

What is the criteria to define whether a PLL is an "analog" PLL or a "digital" one? The general criteria is up to the fact that whether the three blocks (phase detector, loop filter, oscillator) in Figure 1.2 are digital intensively or analog intensively designed. The definition of digital PLL is referring to any PLL architecture with all the three blocks designed in pure digital way or digital intensively way, compared with the conventional analog structure such as charge-pump. As one of the most exciting breakthroughs in RFIC field for the past ten years, ADPLL has founded promising positions in applications from wireline to wireless systems, from Mega Hz to mm-wave products and from utra-low power oriented to high performance intentioned applications. Though shown in numerous patents from US to Europe, described in multitudinous publications from conferences to journals, most of the proposed ADPLL designs could be categorized into two types in terms of the "feedback phase detector" implementation: divider-based and counter-based ADPLLs.

#### **2.1.1.** Divider-based Digital PLL

The divider-based DPLL architecture, shown in Figure 2.1, could be considered as a direct digital equivalent structure of the Charge-Pump PLL [2]. As the name implies, a divider is situated at the first stage of the feedback loop directly after the DCO output to divide down the variable clock CKV before feeding it to TDC. A  $\Sigma\Delta$  modulator based programmable divider determines the output frequency by the ratio set in it. The TDC tracks the phase error by digitizing the difference in the phase of the reference clock and downward scaled output frequency. Thus the digital phase detector consists of a TDC and a programmable divider. According to the digitized phase error info, digital loop filter generates control words to adjust the DCO frequency to track the reference input. Referring to Figure 1.3, conclusion could be drawn that the same operation scheme is shared by both of the charge-pump PLL and this divider based architecture. This divider-based type DPLL is already under investigation from 1980s because of the fact that not too much deviation is there compared with charge-pump version.



Figure 2.1: Divider-based DPLL.

#### 2.1.2. Counter-based Digital PLL

This section introduces an architecture works in real time domain as well as digital domain. This architecture is proposed in the beginning of this century by TI [3] as a breakthrough in PLL design.

One of the breakthroughs in this architecture lies in the oscillator design. The DCO is capacitor array based and thus much more compatible with modern submicron technology nodes compared with conventional VCO design as discussed in [4] [5]. Another important breakthrough lies in the operation scheme of this novel system. It operates in pure digitized phase domain. Figure 2.2 shows a simplified block-diagram of the counter based ADPLL. In the feedback loop there is no divider existing anymore. Instead, the output frequency is set by the frequency command word (FCW). FCW is determined by the supposed number of cycles of output clock within one reference cycle. Hence, accumulating the FCW at reference rate provides the output period normalized reference phase  $\phi_R$ . The integer part of the output phase  $\phi_{CKV}$  is obtained by counting the number of output clock cycles. The fractional phase error is generally quantized by a fractional phase error quantizer (usually realized by TDC). However, the design challenges are generally at the TDC and DCO, which are mixed signal circuits with digital IOs.



Figure 2.2: Counter-based DPLL.

Digitized LF is also an eventful breakthrough in this class of ADPLL. Fully synthesizable implementation not only shrinks significantly the area compared to its analog correspondence, but also makes it realizable to configure LF'S type as well as coefficients dynamically during normal operation without impacting phase error, e.g., gear-shifting technique [4]. Thus much more headroom are allowed to control the closed-loop bandwidth and loop dynamics, benefiting from the reconfigurability brought by the digitalization. Besides, frequency or phase modulation could also be accomplished thanks to the freedom in the FCW control path. Thus, frequency modulation (FM) can be integrated with ADPLL directly which is much simpler compared with conventional structure as shown in Figure 1.3.

#### **2.1.3.** Comparison of Architectures

Based on the two simplified block diagrams shown above, a conclusion could be drawn easily that the main difference of the two structures lies in the feedback loop. This main difference results in distinction in terms of power and phase noise as discussed below.

**1.Divider-based architecture** [6] The essential of the scheme is to first scale down the output frequency to the reference rate and then quantize the phase

Merit	Divider-based	Counter-based
Power	Lower	higher due to the high TDC operation frequency
Complexity	Worse	simpler due to short operation range required
Flexibility	Less	Tricks could be introduced due to operation scheme

Table 2.1: Comparison of the two basic DPLL architectures in general.

difference by TDC. Thus the phase detector is quantizing both the integer and fractional phase error in the same block. In addition, the covering range required for the TDC requires a prohibiting large since the whole reference period has to be covered which is really a disadvantage of this structure. Otherwise, a frequency detection function has to be added. The positive aspect of this structure is the TDC operation frequency is only at reference rate.

**2. Counter-based architecture** The essential of the scheme is based in purely phase domain. Counter and the reference phase accumulator are introduced to quantize the integer part of the phase error. While the TDC is holding the burden of quantizing the fractional phase error. Thus the required operation range for the TDC is only one output frequency period and the counter could be turned off in lockin state since integer phase error is supposed to be zero under that circumstance. Nevertheless, without special clock gating technique, the operation frequency of the TDC has to be DCO output frequency which is usually rather high.

Table 2.1 is a summary conclusion for the comparison in terms of power and complexity as well as flexibility. (This comparison is based on the typical designs. Special techniques such as clock gating, etc. may change the results.)

Based on all the discussion above, counter-based architecture is chosen for this project due to its lower complexity as well as more flexibility.

#### **2.2.** ADPLL System Level Analysis

This section introduces briefly the system level analysis of ADPLL system in terms of frequency response as well as noise sources and their transfer paths in the loop. For more details please refer to [4] [7]. Blocks' impacts for the phase noise profile would be also investigate and summarised.

#### **2.2.1.** PLL Frequency Response

Figure 2.3 shows a general linearised, s-domain model of an ADPLL frequency synthesizer. It is valid as long as the fluctuation frequencies of interest are much smaller than the sampling rate which is the reference  $f_R$  and thus it commonly holds as long as Gardner's rule [7] (i.e., the PLL bandwidth  $f_{BW}$  is at least 10 times smaller than the sampling frequency) holds.

The input and output variables in Figure 2.3 are all digitized phases: reference phase  $\phi_R$ , DCO output phase  $\phi_{CKV}$  and after-divider phase  $\phi_{CKVD}$  as  $\phi_{CKV} = \phi_{CKVD}/M$ . The quantity N is the ratio between the divided DCO clock and FREF and is equivalent to FCW here as

$$N = \frac{f_v/M}{f_R} \tag{2.1}$$



Figure 2.3: Linearised (s-domain) ADPLL.

The block LF(s) shown in Figure 2.3 could be three difference types: it can be a type-I, with only a proportional gain  $\alpha$ , or type-II with both proportional ( $\alpha$ ) and integral ( $\rho$ ) paths, or of higher order with IIR filter turned on. These LF parameters are programmable and can be dynamically configured during normal PLL operation. For the following frequency response analysis, type-II operation would be assumed due to its universality (type-I could be considered as type-II with  $\rho = 0$ ).

For convenience, we would start by open the loop at  $\phi_{CKVD}$  to calculate the

open-loop transfer function  $H_{ol}(s)$  for the type-II case. It is given by

$$H_{ol}(s) = \frac{\phi_{CKVD}}{N * \phi_R}$$
  
=  $(\alpha + \frac{\rho f_R}{s}) \cdot \frac{f_R}{\frac{\Lambda}{K_{DCO}}} \cdot \frac{K_{DCO}}{s} \cdot \frac{1}{M}$  (2.2)

Assuming that the DCO gain is estimated correctly,  $H_o l(s)$  reduces to

$$H_{ol}(s) = \frac{\phi_{CKVD}}{N * \phi_R}$$

$$= (\alpha + \frac{\rho f_R}{s}) \bullet \frac{f_R}{s} \bullet \frac{1}{M}$$
(2.3)

The closed-loop transfer function can then be expressed as

$$H_{cl}(s) = \frac{\phi_{CKV}}{\phi_R}$$
  
=  $\frac{N \cdot M \cdot H_{ol}(s)}{1 + H_{ol}(s)}$   
=  $N \cdot \frac{\alpha f_R s + \rho f_R^2}{s^2 + \frac{\alpha f_R}{M} s + \frac{\rho}{M} f_R^2}$  (2.4)

This can be compared to the classical, two-pole system transfer function

$$H(s) = N \frac{2\xi\omega_n s + \omega_n^2}{s^2 + 2\xi\omega_n s + \omega_n^2}$$
(2.5)

where  $\xi$  is the damping factor and  $\omega_n$  is the non-damped, natural frequency. The zero lies at  $\omega_z = -\omega_n/2\xi$ . According to the analogy between Eq. 2.4 and Eq. 2.5 conclusion could be drawn as,

$$\omega_n = \sqrt{\frac{\rho}{M}} \bullet f_R \tag{2.6}$$

and

$$\xi = \frac{1}{2} \cdot \frac{\alpha}{\sqrt{M\rho}} \tag{2.7}$$

For a type-I loop, the closed-loop transfer function simplifies to

$$H_{cl}(s) = N \bullet \frac{\alpha f_R}{s + \frac{\alpha f_R}{M}}$$
(2.8)

and the 3-dB bandwidth of the loop is  $f_{BW} = \alpha f_R / (2\pi M)$ 

Now, considering the IIR filter in the proportional path of the LF, the transfer function of a one stage IIR filter in s-domain is

$$H_{IIR}(s) = \frac{1 + s/f_R}{1 + s/\lambda f_R}$$
(2.9)

Four independently controlled IIR stages are also implemented as an extra option to shape the phase noise and attenuate the reference and the TDC quantization noise at -80 dB/decade slope. This strong filtering also helps the attenuate the close-in fractional spurs. Each IIR stage has an attenuation factor  $\lambda_i$ , and the open-loop transfer function becomes

$$H_{ol,iir}(s) = (\alpha + \frac{\rho f_R}{s}) \bullet \frac{f_R}{s} \bullet \frac{1}{M} \bullet \prod_{i=0}^3 \frac{1 + s/f_R}{1 + s/\lambda f_R}$$
(2.10)

#### 2.2.2. Noise Sources and Noise Transfer Function

Linear ADPLL model including phase noise sources is shown in Figure 2.4. The  $\phi_{n,R}$  is the phase noise of the input reference clock. Its transfer function is the same as the closed-loop transfer function of Eq. 2.4. The TDC and the DCO are the only two places that noise could be injected to the system internally. Due to its digital nature, the rest of the system is free from time or voltage domain perturbations. The DCO phase noise  $\phi_{n,DCO}$  would experience high-pass filtering by the loop. The closed-loop transfer function is

$$H_{cl,DCO}(s) = \frac{\phi_{CKV}}{\phi_{n,DCO}} = \frac{1}{1 + H_{ol}(s)} = \frac{s^2}{s^2 + \frac{\alpha f_R}{M}s + \frac{\rho}{M}f_R^2}$$
(2.11)

indicating that the DCO noise has a high-pass characteristics and that it dominates the PLL phase noise outside the loop bandwidth. For phase noise at low frequency offsets (i.e., in-band part), the type-II loop will suppress it at a slope of 40 dB/decade.

The second internal noise source  $\phi_{n,FEC}$  arises from the fractional phase error counter (usually accomplished by TDC) operation of calculating  $\epsilon$ . Here use TDC


Figure 2.4: Noise sources in ADPLL.

as example for convenience, it has several components: quantization, linearity, and randomness due to thermal effects, in which the quantization noise governed by Eq. 2.12 is the major contributor [4].

$$\mathcal{L} = \frac{(2\pi)^2}{12} \left(\frac{\Delta t_{res}}{M \bullet T_V}^2\right) \frac{1}{f_R}$$
(2.12)

In Eq. 2.12,  $\Delta t_{res}$  is the time resolution of the TDC, and  $M \cdot T_V$  is the afterdivider clock period at the input of TDC. The factor M is due to the divider. The closed-loop transfer function of the TDC noise can be expressed as

$$H_{cl}(s) = \frac{\phi_{CKV}}{\phi_{n,TDC}}$$
  
=  $\frac{M \bullet H_{ol}(s)}{1 + H_{ol}(s)}$   
=  $\frac{\alpha f_R s + \rho f_R^2}{s^2 + \frac{\alpha f_R}{M} s + \frac{\rho}{M} f_R^2}$  (2.13)

which is a low-pass response with a gain factor of M within the loop bandwidth. Therefore the phase noise at the ADPLL RF output due to TDC quantization noise is simply

$$\mathcal{L} = \frac{(2\pi)^2}{12} (\frac{\Delta t_{res}}{T_V}^2) \frac{1}{f_R}$$
(2.14)

Besides the three mentioned above, the finite frequency resolution of the DCO also contributes to the phase noise at the output and this should be kept much lower than the natural phase noise of the DCO in order to be negligible. Similar to Eq. 2.12, the phase noise due to DCO frequency quantization could be derived [4]

as

$$\mathcal{L} = \frac{1}{12} \left(\frac{\Delta f_{res}}{\Delta f}\right)^2 \frac{1}{f_R} \left(sinc\frac{\Delta f}{f_R}\right)^2 \tag{2.15}$$

Since the DCO input tuning word is held constant between two different values, the white noise assumption is not fully justified. Hence, there is a sinc function in E.q 2.15 to account for the zero-order hold operation on the input tuning word of DCO.

The quantization noise of the DCO has a 20 dB/decade attenuation, similar to that of the up-converted thermal noise from the oscillator. As long as this quantization noise is kept sufficiently low compared to the inherent phase noise of the oscillator, the overall phase noise is not significantly affected. Usually the raw resolution of the DCO is too coarse for the required specification. For that reason,  $\Sigma\Delta$  dithering on the smallest switched capacitance is applied. As a consequence, the quantization noise is shaped in frequency resulting in

$$\mathcal{L} = \frac{1}{12} \left(\frac{\Delta f_{res,eq}}{\Delta f}\right)^2 \frac{1}{f_{dith}} \left(2sinc\frac{\pi \Delta f}{f_{dith}}\right)^{2n}$$
(2.16)

As a consequence, the quantization noise is shaped in frequency resulting as shown in Eq. 2.16, where  $f_{dith}$  is the dithering frequency and n the dithering order. The equivalent frequency resolution after dithering is given as,

$$\Delta f_{res,eq} = \frac{\Delta f_{res}}{2^{W_{TF}}} \tag{2.17}$$

where  $W_{TF}$  is the number of the dithering bits.

The phase noise spectrum at ADPLL RF output is depicted in Figure 2.5, which follows the previous s-domain analysis and all the related parameters settings obey a general case only for convenience here. Above all, together with reviewing the designs in [8] [9] [10], we can already draw the general conclusion for the common metrics of ADPLL's playmakers as summarized in Table 2.2 shown below.

Though bandwidth is determined by the filter characteristics, there is a optimized value under a certain power budget. According to the conclusion in [8], it should be around the position where the open loop DCO phase noise spectrum and loop noise intersects. However, spur issue is more complex and it would be investigated in more details in the next chapter.



Figure 2.5: Phase noise spectrum at the output of ADPLL.

Table 2.2:	Common	metrics	and	depending	blocks
------------	--------	---------	-----	-----------	--------

common metrics	dependence	
Phase noise	in-band (phase detector+reference),outside-band (DCO)	
Bandwidth	Loop Filter	
locking behaviour	Loop filter+ phase detector	
Power	Mainly consumed in DCO and phase detector	

# **2.3.** Worst Inband Fractional Spurious Components of ADPLL Output Spectrum

The term spurious denotes any undesirable signal presents at the output of a frequency system or any portion of it. Here the fractional spur terminology used in ADPLL are defined as those undesired tones lying at multiples of  $FCW_{frac} \cdot f_R$  on both sides of the carrier. Generally fractional spurs are channel depending and the analysis could be rather complicated. These spurs are mainly originated from the feedback phase detector path. Any periodical error introduced from limited res-

olution and mismatch in phase detector under fractional-N mode would modulate the output of the ADPLL. Under the lock-in state, the frequency fluctuation of output is as small as tens of kHz so this error-modulating procedure would be exactly a narrowband FM behaviour. Thanks to the low-pass filtering characteristic loop, the out-of-band fractional spurs generally will not cause serious problem. What is of interest to designers are usually the positions and levels of worst fractional spurs shown in-band. Position estimation theory are mature and discussed in previous publications [11] [12] [13] while the spur level estimation has not been fully analysed. In the following part of this subsection, spur level estimation derivation would be demonstrated, in three parts according to the origins of the spurs: limited resolution, nonlinearity and gain estimation error.

### **2.3.1.** Limited Resolution Introduced Spur

Due to the gate delay and mismatch limitation, a typical inverter based TDC resolution is limited to 10 ps in 40 nm technology. For most application at 2.4 GHz ISM band this is more than enough. However a TDC with resolution of 10 ps is only a 5.3-bit quantizer to cover the period of a 2.4 GHz signal while the fractional part resolution of FCW is much finer than this, e.g., 16-bit width in this project. Eq. 2.12 holds only under the large-signal assumption (spanning multiple quantization levels). When it comes to the case of close integer FCW channel, the fractional phase is increasing with step size much finer than the TDC resolution. Thus there would be a residue error accumulated periodically and tunes the DCO output spectrum. Figure 2.6 shows the example for the case of such a close integer channel under the assumption of a 4-bit TDC and FCW with 6-bit fractional part for convenience (in real case the FCW fractional part is much finer than 6-bit). As can be observed, the periodical residue (highlighted in green) is resulted with a peak amplitude as large as resolution of TDC with a period equals to 4 (4 = $2^{(6-4)}$ )reference cycles. After locking this sawtooth-like residue would behave as a narrowband frequency modulation at the input of the DCO. In order to find the way of estimation of the spur level, analysis from both close-loop and open-loop aspects are conducted as follows:

### a. Open-loop FM analysis

Frequency modulation (FM) is a process of producing a wave whose instantaneous frequency varies as a function of the instantaneous amplitude of a modulating



Figure 2.6: Periodical residue resulted from TDC limited resolution.

wave at a rate given by the frequency of the modulating source. From Figure 2.6, a periodical residue resulted from the rough resolution of TDC could be observed. This periodically behaved residue could be considered as such a modulating source of DCO block. In this way, DCO output could be expressed in the method of a frequency-modulated signal as follows:

$$V_{out}(t) = V_0 \, \cos[\omega_0 t + 2\pi K_{DCO} \int_0^t V_c(\tau) d\tau + \theta_0]$$
(2.18)

Where  $V_c(\tau)$  is the variation at the input of the DCO while  $\theta_0$  is the initial phase.  $K_{DCO}$  is the DCO sensitivity in Hz.

As we can see from the quantization residue ramping shape regarding time, the "modulation" due to the quantization residue could be modelled as a sawtooth narrowband frequency modulation. As far as this paper is concerned, the modulating signal is one of the Fourier Series components of the sawtooth waveform, which is given by:

$$V_e(t) = A_m \cos(2\pi f_m t)$$
 (2.19)

where  $A_m$  is the peak amplitude of the modulating signal from digital phase error while  $f_m$  is the frequency of it. Thus the variation phase of the modulating signal

in DCO output is

$$K_{VCO} \int_{0}^{t} V_{c}(\tau) d\tau = \frac{2\pi K_{DCO} * A_{m}}{2\pi f_{m}} \sin(2\pi f_{m}t)$$
(2.20)

Here we can define the modulation index m as

$$m = \frac{K_{DCO} * A_m}{f_m}$$

$$= \frac{\Delta f_{peak}}{f_m} rad$$
(2.21)

Suppose the initial phase  $\theta_0$  as 0 for convenience. Now we can rewrite 2.18 again as

$$V_{out}(t) = V_0 \cos \left[\omega_0 t + m * \sin(2\pi f_m t)\right]$$
(2.22)

Now with basic trigonometric knowledge we can get

$$V_{out}(t) = V_0 \cos(2\pi f_0 t) * \cos(m * \sin(2\pi f_m t)) - V_0 \sin(2\pi f_0 t) * \sin(m * \sin(2\pi f_m t))$$
(2.23)

Considering the narrowband modulation, m should be far less than  $\frac{\pi}{2}$ . Thus we would have

$$\cos(m * \sin(2\pi f_m t)) \approx 1 \tag{2.24}$$

$$\sin(m * \sin(2\pi f_m t)) \approx m * \sin(2\pi f_m t)$$
(2.25)

Then

$$V_{out}(t) = V_0 \cos(2\pi f_0 t) - V_0 \sin(2\pi f_0 t) * m * \sin(2\pi f_m t)$$
(2.26)

This is equivalent to

$$V_{out}(t) = V_0 \cos (2\pi f_0 t) - 0.5m * V_0 [\cos (2\pi (f_0 + f_m)t) - \cos (2\pi (f_0 - f_m)t)]$$
(2.27)

In this way we can see that the single sideband (SSB) spur to signal ratio could be written in this way:

$$\frac{spur}{signal} = 20\log_{10}(0.5m) = 20\log_{10}(\frac{\Delta f_{peak}}{2 \bullet f_m})$$
(2.28)

Here fm would be

$$f_m = 2^{n_{TDC} - n_f} f_R$$
  

$$f_m = \frac{FCW_{frac} T_v}{t_{res}} f_R$$
(2.29)

where  $f_R$  is the reference frequency,

# **2.3.** Worst Inband Fractional Spurious Components of ADPLL Output Spectrum

### b. close-loop PM analysis

As the modulating residue is a sawtooth waveform, we need to recall the Fourier Series of it here first,

$$x_{sawtooth}(t) = \frac{A}{2} - \frac{A}{\pi} \sum_{n=1}^{\infty} \frac{\sin(2\pi n f t)}{n}$$
(2.30)

A is the amplitude of the waveform in time domain which is  $t_{res}$  here. Now the inband spur level will be investigated first from close-loop aspect. For convenience, we can start from phase modulation (with sinusoidal modulating signal, it makes no difference whether one speaks of phase or frequency deviation because the two are related by the rate of modulation as Eq. 2.21). Thus rewriting Eq. 2.18 as follows,

$$V_{out}(t) = V_0 \, \sin[\omega_0 t + K_p A_m \, \sin(2\pi f_m t)]$$
(2.31)

The maximum phase deviation is

$$\sigma_{\phi} = k_p A_m \ rad \tag{2.32}$$

It is related to frequency modulation by

$$\sigma_{\phi} = m \tag{2.33}$$

Hence the Eq. 2.31 becomes

$$V_{out}(t) = V_0 \sin \left[\omega_0 t + \sigma_\phi \sin \left(2\pi f_m t\right)\right]$$
  
=  $V_0 \sin \left(\omega_0 t\right) \cos \left[\sigma_\phi \sin \left(2\pi f_m t\right)\right] + V_0 \cos \left(\omega_0 t\right) \sin \left[\sigma_\phi \sin \left(2\pi f_m t\right)\right]$   
(2.34)

Under the assumption of lock-in state, the phase deviation is rather smaller than  $\frac{\pi}{2}$ , Eq. 2.34 becomes,

$$V_{out}(t) \approx V_0 \sin(\omega_0 t) * 1 + V_0 \sigma_{\phi} [\cos(\omega_0 t) \sin(2\pi f_m t)]$$

$$\approx V_0 \sin(\omega_0 t) + \frac{V_0 \sigma_{\phi}}{2} (\sin(\omega_0 t + 2\pi f_m t) - \sin(\omega_0 t - 2\pi f_m t))$$
(2.35)

Under the assumption that thermal noise level in the loop is much smaller than the TDC quantization step, the variance of the timing uncertainty would be as large as

$$\sigma_t^2 = \left(\frac{1}{\pi} \Delta t_{res}\right)^2 \tag{2.36}$$

The equivalent phase noise energy could be estimated as,

$$\sigma_{\phi} = 2\pi \frac{\sigma_t}{T_v} \tag{2.37}$$

where the  $T_{v}$  is one period of the DCO output.

Instead of being spread normally over the span from DC to Nyquist frequency, (i.e., half of the reference frequency  $f_R$ ), the phase noise energy is mainly concentrated at the fractional spur position thus there is no need to normalize the noise energy by the Nyquist frequency span to calculate the spur's level. Hereby the estimated spur level could be expressed in SSB form as,

$$\frac{spur}{signal} = 20 \log_{10}(0.5\sigma_{\phi}) = 20 \log_{10}(2\pi \frac{\frac{1}{\pi} \Delta t_{res}}{2T_{v}})$$

$$= 20 \log_{10}(\frac{t_{res}}{T_{v}})$$
(2.38)

### c. comparison

First, since we have the relation between time domain and frequency domain as

$$\alpha \frac{t_{res}}{T_v} = \frac{f_{res}}{f_R}$$
(2.39)

The corresponding peak deviation frequency is

$$\Delta f_{peak} = \frac{1}{\pi} f_R \bullet \frac{t_{res}}{T_v}$$
(2.40)

Thus Eq.2.28 is rewritten as,

$$\frac{spur}{signal} = 20\log_{10}(\frac{t_{res} * \alpha * f_R}{2\pi T_v * f_m})$$
(2.41)

To show the equivalence between Eq. 2.41 and Eq. 2.38, let's assume there is one fractional spur locates at PLL cutoff frequency  $f_{BW}$  as

$$f_m = f_{BW} = \frac{1}{2\pi} \alpha f_R \tag{2.42}$$

where the tangential open-loop response and closed-loop response are both at unity (actually they are both at -3 dB gain):

$$\frac{spur}{signal} = 20 \log_{10}\left(\frac{t_{res} * \alpha * f_R}{2\pi T_v * f_m}\right)$$

$$= 20 \log_{10}\left(\frac{t_{res}}{T_v}\right)$$
(2.43)

resolution	Estimated (dB)	Simulated (dB)
15 ps	-34.8911	-35.12
64 ps	-22.2893	-22.31

Table 2.3: Validation about close integer spur estimation with FCW= 38.0001.

This is exactly the same as Eq. 2.38. Theoretically estimated results and Verilog simulated results comparisons are summarised in Table 2.3 for the worst inband fractional spur level. No TDC nonlinearity is added into the Verilog model and the channel is chosen to be close integer one (FCW= 38.0001). Bandwidth is set to be 200 kHz while the standard deviation of the jitter from reference clock is 3 ps which is much lower than the TDC resolution.  $T_v$  is 1.2 GHz in this example. However this matching result holds under the condition that the thermal noise level from the loop at the input of the TDC is much lower than the TDC quantization residue level, which is expected. Once they become comparable to each other, the random noise could dither the quantization error and the spurs energy would be flatten around more into the while noise floor. And this could be a solution to reduce the fractional spur as shown in [14] by adding DTC to dither the reference clock before sending it into the TDC.

# **2.3.2.** TDC Nonlinearity Introduced Spur

Unfortunately, the worst case fractional spurs for most of the channels are not resulted from resolution related scheme, but caused by the nonlinearity of TDC. As shown in Figure 2.7, the TDC resolution is set to be infinite but with a nonlinearity. The INL is assumed to be sinusoidal here for convenience but this INL shape would be rather complex in the real implementation and the spur level estimation would be hard to do.

Now the fundamental spur position  $f_{frac}$  is related to the fractional part of FCW as  $f_{frac} = FCW_{frac} \cdot Fref$ . For example in the case with FCW= 38.0001 together with reference clock at 64 MHz, fractional spurs resulted from nonlinearity are anticipated at  $f_{spur}(i) = i \cdot f_{frac}$ , where  $i = (1, 2, \cdots)$  is the index of the spur tones. Generally the ones lies within bandwidth would cause problems since they are not filtered. Nevertheless, the nonlinearity resulted error is still a periodical series, though not sawtooth alike one, the fundamental item of its corresponding



Figure 2.7: Periodical residue from TDC nonlinearity.

Fourier Series can always be expressed in the form of Eq. 2.19. Then the sinusoidal modulated DCO could be expressed by a Bessel function [15] series with modulation index m as defined previously.

$$V_{out}(t) = V_0 \{J_0(m) \sin(\omega_0 t) + J_1(m) [\sin(\omega_0 t + 2\pi f_m t) - \sin(\omega_0 t - 2\pi f_m t)] + J_2(m) [\sin(\omega_0 t + 2\pi 2f_m t) + \sin(\omega_0 t - 2\pi 2f_m t)] + J_3(m) [\sin(\omega_0 t + 2\pi 3f_m t) - \sin(\omega_0 t - 2\pi 3f_m t)] + \cdots \}$$
(2.44)

Only when modulation index m is much smaller than 1, we can have the approximation for the fundamental tone in the form of Eq. 2.38. Under this assumption  $J_1(m) \approx \frac{m}{2}$  while the higher order item would be zero. Regarding m, the peak deviation value is also related to the shape of the INL. As demonstrated in Figure 2.7, assume the max INL is  $t_{INL}$  then we can go through the same procedure discussed before to get a result as

$$\frac{spur}{signal} = 20\log_{10}(\pi \frac{t_{INL}}{T_v})$$
(2.45)

This nonlinearity is also modelled and simulated with a TDC resolution as 0.5 ps which is smaller than the thermal noise in the loop so the quantization residue

could be fully randomized by the thermal noise. Results match with deviation less than 2 dB.

# 2.3.3. TDC Normalization Introduced Spur

When the fractional phase exceeds  $2\pi$ , it wraps back to zero and then ramps step by step again as shown in the Figure 2.8. If the TDC normalization is not done correctly, a residue error would be generated from this non  $2\pi$  normalization and thus introduces a sawtooth alike error tuning the DCO as long as the loop is slow enough compared to the variation of the error. The formula to estimate this kind of spur level shares the same essential derivation as the TDC resolution resulted spur. The level estimation would be complex since the residue depends on the fractional part of FCW. However, once the fractional reference phase's ramping step is smaller than TDC resolution, the situation becomes as simple as shown in Figure 2.8.



Figure 2.8: Periodical residue from wrongly TDC normalization.

Now the spur position would be related to the number of the TDC bits while the level could be expressed as

$$\frac{spur}{signal} = 20\log_{10}(\frac{t_{residue}}{T_v})$$
(2.46)

Sources	Position	Level	Solution
Resolution	$\frac{FCW_{frac}T_v}{t_{res}}f_R$	$20\log_{10}(\frac{t_{res}}{T_v})$	finer resolution
Nonlinearity	$FCW_{frac} * f_R$	$20\log_{10}(\pi\frac{t_{INL}}{T_v})$	better linearity
Normalization	$FCW_{frac} * f_R$	$20\log_{10}(\frac{t_{residue}}{T_v})$	gain calibration

Table 2.4: Fractional Spur Estimation Summary in General

<sup>1</sup> This normalization related error estimation depends on fractional bits of FCW.

Where the residue is related to the TDC bit number n as

$$t_{residue} = 2^{n} * \frac{Gain_{wrong} - Gain_{ideal}}{Gain_{ideal}} \bullet t_{res}$$
(2.47)

The fundamental place of the fractional spurs would be  $f_{frac} = FCW_{frac} * f_R$  since this residue would always wrap back to zero together with fractional reference phase.

Analysis results are summarised in Table 2.4. To draw a conclusion, we need a quantizer with not only good resolution but also good linearity, together with correct gain calibration to make the fractional spur as small as possible.

# **2.4.** Specification of the targeted ADPLL Design

This ultra-low power ADPLL project is dedicated to Bluetooth low energy (BLE) application which is an extension of the conventional Bluetooth standard published with the version 4.0 of the latter in late 2009. Compared with conventional Bluetooth basic the energy demand is required to be reduced by 90% in BLE. This is made possible by relaxing the radio specifications to help saving more power in physical layer. In protocol level, connection set-up is also accelerated to make duty cycle more efficient. For example, the requirements to block interferers from adjacent channels is reduced. Table 2.5 shows the difference between these two versions of Bluetooth standard.

For a Bluetooth transmitter application, a frequency synthesizer can be characterized by the phase noise, spurious tones and switching time. The derivations of specifications are described as follows.

Parameter	Bluetooth basic rate	BLE
RF channels	f=2402+k MHz	f=2402+2k MHz
	k=0,1,,78	k=0,1,,39
Carrier frequency rate	± 75 kHz	$\pm$ 150 kHz
Spurious emissions		
2 MHz	-20 dBm	-20 dBm
3 MHz	-40 dBm	-30 dBm
Required sensitivity	-70 dBm	-70 dBm
Carrier-to interference ratio(CIR)		
1 MHz	0 dB	15 dB
2 MHz	-30 dB	-17 dB
3 MHz	-40 dB	-27 dB

Table 2.5: Bluetooth radio specifications for the basic rate and for the BLE extension

# 2.4.1. Phase Noise Specification

To calculate the tolerable phase noise level, a required signal-to-noise ratio (SNR) has to be assumed in terms of demodulation performance. A GFSK demodulators for a modulation index of h = 0.5 requires an SNR of about 11 dB. To allow for sufficient implementation margin, an SNR of 15 dB is required. Assuming a constant phase noise within the channel bandwidth then the required in-channel average phase noise  $\mathcal{L}_{inband}$  500 kHz bandwidth channel simplifiers to less than -75 dBc/Hz ( $\frac{1}{SNR \cdot BW}$ ) which is really not stringent for most nowadays PLL designs. For spot phase noise requirement at larger offset from the carrier, the adjacent channel Carrier-to-interference ratio has to be taken into consideration which is -17 dB in BLE at 2MHz and this results in a requirement for -92 dBc/Hz level at 2 MHz offset.

Here the specifications of the in-band noise floor is set as -85 dBc/Hz and spot phase noise of -105 dBc/ Hz at 1 MHz offset for much more margin according to the plan of NXP.

# 2.4.2. Spurious Tone Specification

The tolerable spur level according to the spurious emission specification is -20 dBc or -30 dBc at 2MHz or 3MHz offset respectively. The tolerable spur level normalized to the carrier ( $P_{spur}/P_{carrier}$ ) can be derived by considering how much an interferer at a channel offset  $f_{ch}$  has to be suppressed to fall below a certain signal to noise ratio. Thus a target for reference spur less than -70 dB will be enough while inband fractional spurs lower than -32 dB has to be met to avoid serious adjacent channel interference according to NXP's requirement. The fractional spur reduction is also one of the priority of this thesis work.

# 2.4.3. Settling Time Specification

BLE adopts a frequency hop scheme. A frequency hop transceiver is applied to combat interference and fading. The nominal hop rate is 1600 hops/s. A settling time less than 65  $\mu$ s with a accuracy finer than  $\pm$  150 kHz is enough for this target.

Table 2.6 summarizes the targeted specifications according to the above analysis. Compared with current leading designs, the difficulties as well as potential breakthroughs lie in the power consumption and fractional spur level reduction. The trade-off between power and integrated jitter would be pushed to make this ADPLL design work power efficiently, which means to generate as low as possible integrated jitter at a certain low power budget (< 800  $\mu$ W).

# **2.5.** Implementation from System level

## **2.5.1.** Reference Design Introduction

The reference design becomes available to this thesis work and is published in [1]. Here the design concept would be covered briefly.

Though TDC is a mature and exciting topic to explore in the direction of better resolution at lower power consumption, it is already proved there is no certain trade-off between resolution and power in its DPLL application. Besides being used in divider-based DPLL, DTC is already introduced in [16] for fractional divider noise cancellation and again in [17] [18] it is introduced for phase predication in counter-based DPLL. Without a DTC, the TDC's burden in the counter-based structure is to detect the fractional phase error as explained in [4],

$$\varepsilon = 1 - \frac{t_v - t_r}{T_v} \tag{2.48}$$

Technology	GlobalFoundries's 40 nm-LP
Supply Voltage	0.8 V $\sim$ 1 V (separate low supply for DCO)
Power Consumption	< 800 µW
locking range	2.2 GHz~ 3 GHz
Settling time	<60 µs
Reference noise floor	-135 dBc/Hz
Reference spur	<-70 dB
Inband phase noise floor	<-85 dBc/Hz
Phase noise @ 1MHz offset	<-105 dBc/Hz
Integrated RMS jitter	<1.5 ps
Worst fractional spur	<-30 dB

Table 2.6: Summary of Specifications and Conditions

where  $t_r$  is the rising edge of reference clock and  $t_v$  is the rising edge of the feedback variable clock,  $T_v$  is the period of the feedback variable clock,  $\varepsilon$  is the quantization target of TDC. In lock-in state,

$$R_{R,f} + \varepsilon = 1 \tag{2.49}$$

where  $R_{R,f}$  is the fractional part of the reference phase. However it could be expressed now also as,

$$R_{R,f} + 1 - \frac{t_v - t_r}{T_v} = 1$$

$$t_r + (1 - R_{R,f}) * T_v = t_v + T_v$$
(2.50)

Thus under the lock-in state assumption, what TDC is tracking is  $(1 - R_{R,f})$  plus additional noise and residue in the loop, which is a predicable variant  $(1 - R_{R,f})$ plus small turbulence. According to Eq. 2.50, DTC could be introduced to handle the phase prediction work and thus reference clock is always delayed to be aligned to the next rising edge of the feedback variable phase. TDC's work now is to only quantize the residue between these two phases. Figure 2.10 shows the phase prediction diagram. This phase prediction scheme is already implemented in [17] and [1].



Figure 2.9: IMEC's Design Diagram [1].



Figure 2.10: Phase Prediction Scheme.

As shown in Figure 2.9, phase detector in reference design consists of 64 stages of 30 ps DTC followed by a 16 stages of 15 ps TDC to quantize the residue generated from phase prediction. This structure relaxes the power hungry TDC a lot. Together with the smart clock gating technique which is called "snapshot" scheme in [1], only 860  $\mu$  W is consumed to meet the specifications.

To sum up, this phase prediction assisted counter-based DPLL is the best candidate for low power design and thus the phase prediction scheme as well as the clock gating concept are chosen as the starting concept of this thesis design.

# **2.5.2.** DTC-based Phase Detector

Phase detector and DCO are generally where the power is consumed. However, for the oscillator there is always a hard trade-off between power and phase noise [19] [20]. For phase detector, the trade-off is not so clear and it is largely depending on the operation scheme of the phase detector. This lends us much more headroom to design a **power efficient** ADPLL, which means a ADPLL offering **low noise** and **small spur** at **low power**. The reference design brought a really breakthrough in terms of the power saving. However, the advantage of this phase prediction scheme has not been fully taken advantage of and the worst case fractional spur could be as high as -20 dB around as shown in Figure 2.11.

In order to fully explore the advantage of this DTC-based phase detector, analysis has to be done regarding the in-band phase noise level, fractional spur level.Nevertheless, other than the suspicious sources like supply coupling and supply noise which may add spurs via Amplitude Modulation scheme, the main reasons for this fractional spurs are still from the three schemes mentioned above as:

#### a. Fractional Spur and In-band Phase Noise Floor

Essentially, equivalent resolution of DTC-based phase detector is determined by the one with much finer resolution between DTC and TDC. Its operation could be analog to a two stage TDC. DTC is also doing a "quantizing" (phase prediction) and the residue is left to TDC for further quantization, as shown in Figure 2.12. This would be analysed in three cases in terms of the relation between their resolutions:

**I. DTC resolution** » **TDC resolution** Under this circumstance, the DTC limited resolution contributes nothing to the output spur or phase noise. Because the residue error from phase tracking (the fractional phase steps could not be tracked exactly by DTC resolution) would be detected by the finer TDC. Besides, the large residue left from DTC could fully scramble the TDC output thus the quantization noise floor's assumption still holds (determined by TDC) [4]. Only under close-integer case would TDC face the deadzone issue [9]. Under this case. the phase prediction scheme experiences no big difference compared with the TDC-only structure. The only potential impact from DTC stage would be its nonlinearity and gain estimation error, which could manipulate the TDC output and hence further modulates the output spectrum. Other than these, DTC is only assisting the narrow-range TDC.



Figure 2.11: Measured worst case spur in IMEC design

**II. DTC resolution**  $\approx$  **TDC resolution** This is the case when DTC resolution is slightly larger or smaller than TDC resolution. Under this circumstance, what DTC could not track can not be detected out by TDC also. Hereby the TDC is always facing the deadzone problem theoretically since DTC always aligns the two input to the extent that the difference is smaller than TDC's resolution. Nevertheless, in real case the strong nonlinearity of DTC and thermal noise could enlarge the difference

and relax this serious problem. To solve this problem, what can be done is either to add dithering to reference or to use worse crystal to save the TDC out of deadzone.



Figure 2.12: DTC working principal.

**III. DTC resolution « TDC resolution** In this situation, what DTC does is exactly to align the input of TDC into close integer case and it would take extremely long time for the accumulated error from DTC phase tracking to be detected by TDC. Under this case the spur and in-band noise floor would be dominated by TDC mainly.

In a word, in terms of resolution, as long as TDC's could recognise the residue left from DTC, this equivalent resolution of phase error detection is determined by TDC mainly. In terms of nonlinearity resulted spur, both of DTC and TDC would contribute and generally it is DTC dominating since DTC's transfer characteristic would be fully travelled while fractional phase is ramping periodically.

Besides, a wrongly estimated DTC gain also contributes significantly to the fractional spur. Since DTC's gain would determine the input error of the TDC, DTC gain estimation is not only important for spurs but also important for locking. It would cause serious spur problem in close integer channel with wrongly estimated DTC gain. Nevertheless, this problem can be totally solved by LMS algorithm as

depicted in [17].

The reference design measured results is rebuilt in the model to verify the analysis mentioned above, and the result is shown in Figure 2.14. The reference noise floor is -120 dBc/Hz and  $\alpha = 2^{-5}$  while  $\rho = 2^{-9}$ . Channel is chosen as FCW=38.001 and reference clock is 32 MHz with RBW=1 kHz. DTC's resolution is 22 ps while TDC's resolution is 15 ps. All these parameters are set up as done during the measurement. Nonlinearity resulted fractional spurs are concentrating at multiple of 32 kHz. The INL of DTC is plot in Figure 2.13. The simulation result matches the analysis really well in the sense that:

**Inband phase noise** Theoretically the inband phase noise is determined by TDC resolution and this is verified as seen in figure.

**Nonliearity Spur** This is dominated by DTC and simulated result is -25.74 dBc/Hz (-55.74+10log(RBW)). Estimated level is -25.89 dBc/Hz with 0.6 LSB INL (shown in Figure 2.13).

**Quantization related spur** The stronger spur outside bandwidth at 1.188 MHz is resulted from quantization of DTC's resolution. (This is the case in which DTC is slightly more coarse than TDC and the residue is left). This position is also similar to estimated result which is 1.19 MHz. The position is almost 10 times the bandwidth thus the equivalent inband quantization spur level is -33.33 dBc/Hz (-83.33+10log (RBW)+20). The estimated level is -31.39 dBc/Hz. However this 2 dB deviation could be explained since we roughly add the 20 dB compensation for the loop filtering effect.

### b. DTC-based Bang-Bang Phase Detector

Above all, the analysis done for spur estimation in section 2.3 still applies to the DTC-TDC DPLL since DTC-TDC structure could be considered as an alternative to two stage TDC. The only difference is that the DTC is always aligning the input signals of the succeeding TDC and this causes deadzone problem in potential. However, an integer alike operation scheme implies us to use Bang-Bang phase detector instead of TDC in lock-in state since Bang-Bang based DPLL is well known for good phase noise performance in integer-N mode compared with TDC [21] [22]. Instead of a mid-tread alike TDC, Bang-Bang's transfer function is rather nonlinear with only +1 or -1 output as could be seen in Figure 2.15. As a 1-bit mid-rise quantizer, it



Figure 2.13: Nonlinearity of the rebuilt DTC.



Figure 2.14: Reference design rebuilt.

2



Figure 2.15: (a)TDC as mid-tread quantizer (b) Bang-Bang as mid-rise quantizer

could be as simple as a D flip-flop. The deadzone (TDC outputs zero while error is accumulating) issue in close integer channel for TDC is not a problem anymore for Bang-Bang based DPLL. However, without random noise to dither the input signal, Bang-Bang's strong nonlinearity would cause limit cycle problem also and make the loop hard to be analysed in a linear s-domain model.

Thanks to the thermal noise in the loop which is mainly determined by reference clock, the inputs would be randomized and the transfer function is shaped by the thermal noise as shown in Figure 2.16.

The transfer function of the Bang-Bang phase detector is then linearised by the noise with a equivalent gain as

$$K_{BangBang} = \sqrt{\frac{2}{\pi} \frac{1}{\sigma_t}}$$
(2.51)

where  $\sigma_t$  is the standard deviation of the thermal noise in the loop under assumption that the noise is Gaussian distributed. Now in the closed loop transfer function the gain of Bang-Bang would be observed. Use type-I close loop function as an example which should be,

$$H_{cl}(s) = N \bullet \frac{\alpha K_{BangBang} f_R}{s + \frac{\alpha K_{BangBang} f_R}{M}}$$
(2.52)

and the 3-dB bandwidth of the loop is  $f_{BW} = \alpha K_{BangBang} f_R / (2\pi M)$  now.

By far, two negative issues with Bang-Bang phase detector have to be clarified here. First is the input phase difference should not be too large compared with



Figure 2.16: (a)Simplified Bang-Bang (b)Input edge shifted by thermal noise (c) Gain linearised by thermal noise

thermal noise in the loop otherwise the transfer function is still strongly nonlinear. Another drawback of Bang-Bang phase detector is the loop bandwidth will be impacted by the gain of the phase detector which means the bandwidth would be reference noise depending. However since the quality of the reference clock could be ensured with no big variance, it is expected to see the bandwidth maintained stable.

Back to the first issue, a rather fine DTC with acceptable linearity is required to ensure the Bang-Bang phase detector's input is always randomized by the loop thermal noise. In such a way the "TDC" is always working in close integer mode, ensured by a good DTC. Instead of a mid-tread quantizer TDC, mid-rise based Bang-Bang phase detector is adopted to take advantage of the thermal noise so that the conventional deadzone problem is avoided. Now all of the phase noise as well as fractional spur issue are determined by DTC. No dithering is needed at all and in theory the PLL is expected to work perfectly as Bang-Bang DPLL does under integer-N mode as long as the DTC is extremely linear with picosecond resolution. Hence, all of the burden of the TDC are transferred into DTC and there are good reasons to do so in terms of power, mismatch as well as compatibility with digital calibration ( under the target of ultra-low power, delay chain based DTC or TDC is preferred to maintain low power as well as simplicity): **In terms of power** DTC does not need the D Flip-Flop (DFF) to sample the output value and is consuming lower power due to simplicity compared with TDC.

**In terms of linearity** If the delay chains for both TDC and DTC are the same, TDC will get additional mismatch contributed from the array of DFFs.

**In terms of calibration** DTC is a structure like DAC to whom digital techniques such as predistortion as well as DEM can be applied easily. On the other hand, it would be much more power hungry to apply those tricks to TDC.

Another good reason to go for DTC in terms of resolution would be covered in Chapter 4. Yet DTC-only-based phase detector has its own drawback also, due to the limited linear detection range of Bang-Bang quantizer, it is easier to lose lock. Besides, it is undesired to see the bandwidth may be impacted by the noise in the loop.

Driven by all the motivations mentioned above, the final structure is proposed as consisting of a DTC offering resolution at thermal noise level (which is 2 ps around corresponding to a reference clock of -135 dBc/Hz noise floor) with good linearity and a Bang-Bang detector. DEM is also implemented to relax potential nonlinearity issue.

### **2.5.3.** Proposed System

The structure proposed is shown in Figure 2.17. The "CDTC" means the coarse DTC here while "FDTC" represents the fine stage.

The phase detector is based on the discussion above. A coarse-fine DTC structure is adopted here for power saving and will be fully explained in Chapter 4. The 16-stage coarse DTC unit cell is implemented with a resolution of 58 ps to cover more than the required range for safety. Then 32-stage 2 ps DTC unit cell is designed to make the phase detector's resolution comparable with thermal noise to fully random the Bang-Bang detector's nonlinear transfer function. DEM is applied to the DTC part to relax the nonlinearity issue. An improved clock gating technique called "time freezer" is proposed here to generate the re-timed reference clock as well as the variable input to the Bang-Bang detector.

This design is taped out on Sep. 11th. Two additional techniques are not taped out due to the limit of time, they are:

1. Quadrature phase assisted DTC. This is done by taking advantage of the

4 phase output from the divider by 2. By choosing the phase which is closest to reference edge from the four phases generated, the required range for DTC is reduced to only  $\frac{1}{4}$  of the original  $T_v$ .

2. TDC assisted fast locking. Full range TDC has a advantage which is to let the loop always observe a linearly quantized phase error and thus locking would be rather fast. Before lock-in, DTC is not doing the correct phase prediction and this results in a drawback of slower settling time compared to full range TDC based ADPLL designs. However, a Full range TDC could help observe the fractional phase error in the range of full  $T_v$  period at the beginning. The TDC could be switched to a 1 bit Bang-Bang operation after lock-in. Since spurious and power as well as phase noise are discussed only after the loop locks, the resolution, linearity of the full range TDC has no impact at all. In this way, the advantages of both TDC and Bang-Bang are combined together.



Figure 2.17: Simplified diagram of proposed architecture

All of these would be demonstrated in detail in Chapter 4.

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# 3

# ADPLL Modelling and Simulation

Jitter or phase noise is always of great concern to the designers of PLL system. Jitter and phase noise are different ways of describing an undesired variation in the timing of events at the output of the PLL. They are difficult to predict with traditional circuit simulators because the PLL generates repetitive switching events as an essential part of its operation.

Direct simulation such as SpectreRF based simulation is accurate but usually takes time as long as several months to get characterized smooth in-band phase noise and spurs. Based on the analysis work done in Chapter 2, a more time efficient as well as accurate method is proposed to model and simulate the ADPLL. In this chapter, the modelling and simulation methodology of ADPLL is illustrated.

# 3.1. General Methods

G enerally there are two different ways of simulating a PLL system: the voltage domain direct simulation and time domain event-driven based simulation are discussed as follows:

# 3.1.1. Direct Simulation

In many circumstances, SpectreRF can be directly applied to predict the noise performance of a PLL. Nevertheless the precondition is that the PLL must at least have a periodic steady state solution. This rules out systems such as Bang-Bang clock and data recovery circuits and fractional-N synthesizers because they behave in a chaotic way by design. It also rules out any PLL that is implemented with a phase detector that has a dead zone. A dead zone has the effect of opening the loop and letting the phase drift seemingly at random when the phase of the reference and the output of the voltage-controlled oscillator (VCO) are close. This gives these PLLs a chaotic nature. To perform a noise analysis, SpectreRF must first compute the steady-state solution of the circuit with its periodic steady state (PSS) analysis. If the PLL does not have a periodic solution then simulation will not converge. Unfortunately both the Bang-Bang DPLL and TDC based ADPLL in close integer mode will have the problem mentioned above. Besides, the oscillator is working at the frequency of several GHz or tens of GHz while the frequency accuracy requirement for the system is tens or hundreds of Hz. To get the response of the oscillator with enough accuracy, simulator will execute the simulation in step of ps or fs. On the other hand, to characterize an accurate and smooth in-band phase noise the simulation needs to take several ms which would require simulation time as long as several months. Together with thousands of transistors involved, the extremely long time require by SpectreRF simulation for phase noise estimation would be too long to be adopted.

Nevertheless SpectreRF simulation can still provide an accurate estimation of the design in terms of power, phase noise in transistor level. Even though PSS is not applicative for some cases, VCO output time stamps or input tuning signal information could still be adopted for processing to estimate the phase noise. Besides, the meaningful side of SpectreRF simulation is that the power consumption could be estimated roughly.

### **3.1.2.** Event Driven Verilog-based Simulation

[1] gives a hint on event driven simulation method and this Verilog/VHDL based simulation is already widely used and verified among DPLL designers. What's more, the time domain operation scheme of the counter-based ADPLL makes itself a perfect candidate for event driven Verilog-based simulation. The phase noise and locking behaviour could be predicted really well thanks to the modern CMOS processes

Metric	SpectreRF	Event Driven Verilog-based	
Power	accurate	only possible via digital flow	
		to estimate netlist file at a rough accuracy	
Locking Behaviour	accurate	accurate	
Phase Noise	accurate	accurate	
Spurious Tones	accurate	accurate	
Time needed	5 days for 50 $\mu$ s	10 minutes for 1 ms	

Table 3.1: Methodologies Comparison

which are offering steep timing stamps. Table 3.1 summarises the differences between the two methodologies mentioned above.

# **3.2.** Mixed Signal Simulation Methodology

Though SpectreRF based voltage domain direct simulation could give us sufficient information of what we want to know about the design, time required for this simulation is prohibiting long. Instead of wasting time in waiting several months for the simulated results, a sufficient but more efficient simulation method is anticipated to try different ideas and structures by simulating the model within relatively short time.

Table 3.2 shows the proposed mixed signal simulation methodology which is used in this thesis design. All the design results presented in this thesis are based on this flow. One thing has to be explained here is about the AMS in the step 4 in Table 3.2. This is a simulator available in Cadence which is a mixed signal simulator. It combines the Verilog-based script together with schematic netlist in one simulation. The essential principal is to simulate all the schematic netlist based parts by SpectreRF while simulating the digital Verilog codes by NCSim. The interaction between digital code and voltage domain analog netlist is finished by the predefined rule guided conversion between analog and digital domain.

Step	Phase Detector (Mixed)	DCO (RF)	LF (Digital)
1	Matlab	Matlab	Matlab
2	Verilog	Verilog	Verilog
3(SpectreRF)	Sch	Sch	Synthesized Sch
4(AMS)	Sch	Verilog (sch)	Synthesized netlist
5(NCsim)	Verilog(lay)	Verilog(lay)	with backend timing annotation

Table 3.2: Proposed Methodology

<sup>1</sup> Sch is shot for Schematic here.

<sup>2</sup> Lay is short for post-layout extracted information here.

# **3.3.** Phase Detector Modelling

As depicted in Chapter 2, the phase detector realized in this thesis consists of DTC, Bang-Bang detector and the architecture is mainly inverter based as would be explained in Chapter 4. This is really easy to be modelled in Verilog while the nonidealities are considered as follow:

### **3.3.1.** Mismatch

Mismatch among different delay stages would cause serious spurious problem and this mismatch is modelled by a Gaussian distribution model. However, for a certain INL shape we can also directly predefined the delay for each stage.

# 3.3.2. Accumulated Noise

The loop noise in the reference path is mainly from reference clock jitter and the accumulated jitter from all the delay stages of the DTC and Bang-Bang detector.

Since these noise sources are confronting the same transfer function, all of them can be referred to the noise in the reference clock. The reference clock used in the lab is with the noise floor in the range from -140 dBc/Hz to -135 dBc/Hz.

### **3.3.3.** Supply Noise and Dirty Ground

The supply noise and ground bouncing issue is rather complex and may bring global delay variation to interfere the loop locking. However this is a phenomenon complex to model but won't impact the locking. Corresponding solution is proposed in Chapter 4.

### 3.3.4. Metastability

Metastability could be an issue for Bang-Bang phase detector. It adds really ignorable effect to the result in terms of the relatively smaller metastability window compared with the thermal noise in the loop. In this way the transfer function of the Bang-Bang detector is not so much impacted by this effect. Besides (as will be explained later) in the retiming block, CKV is resampled twice to create a 1.6 ns separation between variable input (CKVG) to the Bang-Bang and resampled reference CKR. Thus time is rather sufficient for the 1 bit quantizer's output to reach to a well defined level.

# 3.4. RF Modelling

# 3.4.1. DCO

The oscillator jitter part and wander part are modelled according to [2]. However, the flicker noise part mainly impacts the integrated jitter but won't impact the locking behaviour at all. Besides, considering the flicker noise is not a serious problem in terms of the design specification, this part is not included in my model. For jitter noise, the standard deviation of the time domain Gaussian distributed jitter is

$$\sigma_{\Delta t} = \frac{T_0}{2\pi} \sqrt{\mathcal{L}f_0} \tag{3.1}$$

where  $\ensuremath{\mathcal{L}}$  is the noise floor and for wander noise the standard deviation is

$$\sigma_{\Delta t} = \frac{\Delta f}{f_0} \sqrt{T_0} \sqrt{\mathcal{L}(\Delta f)}$$
(3.2)

where  $\Delta f$  represents the offset frequency from the output carrier. The noise is added to the DCO output in time domain based on Gaussian distribution model.

# 3.4.2. Feedback Divider

The feedback divider impacts the output phase noise in the same way as the DCO does. Thus they share the same modelling methodology. The phase noise profile could be easily obtained together with DCO from post-layout simulation.

# **3.5.** Digital Flow

The digital flow is completed in two steps. First is to translate RTL code to transistor based netlist with the help of RTL Compiler. Then Encounter is needed for the back-end flow to generate the corresponding layout.

Timing information of gate delay can be extracted from the digital flow. This can be added to the Verilog-based model easily for more accuracy without increasing simulation time significantly. The digital part mainly impacts the locking while contributes almost nothing to the output phase noise spectrum.

# **3.6.** Verification of the Model

Measurement of the chip in [3] is conducted during the thesis. Thus the model based simulation results and the corresponding measurement results can be compared for verification in terms of worst case in-band fractional spurs and the phase noise. Comparisons are shown as follows:

In terms of settling time (please refer to Figure 3.1 and Figure 3.2), not only the Bang-Bang behaviour and type II behaviour could be rebuilt but also the settling time is more or less the same as 16  $\mu$ s. The deviation comes from the difference in initial phase offset between reference clock and variable output which is not known and can't be rebuilt.

In terms of the phase noise and fractional spur plot (please refer to Figure 3.3 and Figure 3.4), FCW=38.0001 is used here with -120 dBc/Hz 32 MHz reference. All the other parameters are also set up in the same way other than linearity, which is modelled roughly based on measured level. The noise floor of -82 dBc/Hz and a peak at -76 dBc/Hz match the measured results. Besides, the spur positions (resolution resulted at 111 kHz and nonlinearity fundamental at 3 kHz) are also rebuilt. However, the fractional spurs levels due to nonlinearity are hard to rebuilt but the quantization resulted one matches the measurement result as -26 dBc/Hz and the worst case of fractional spurs could easily reach the measured worst case with INL increased. Thus the simulation method is verified to be reliable for estimation.



Figure 3.1: Measured settling behaviour.



Figure 3.2: Simulated settling behaviour.



Figure 3.3: Measured phase noise plot.



Figure 3.4: Simulated phase noise plot.
# References

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# 4

# ADPLL Implementation in Transistor-Level

In this chapter, the whole design of an ultra-low power ADPLL for BLE application is presented in three parts. Section 4.1 demonstrates the digital flow based synthesizable low speed digital logic blocks. Section 4.2 describes the mixed signal blocks such as DTC and counter design. The DCO and divider design which belongs to RF domain is covered in Section 4.3.

# 4.1. Low Speed Digital Implementation

igure 4.1 shows a simplified top-level block diagram of the low speed digital logic. Digital flow in specification (RTL code) level is based on Verilog due to its compatibility with NXP digital flow.

The low speed logics are all in the resampled reference clock (CKR) domain which contains a finite state machine, interface to tx data, phase detector, and a loop filter for each of the DCO's three banks: PVT, acquisition, and tracking. The input frequency control word (FCW) is used to generate the required frequency according to Eq. 4.1.

$$f_{outdiv} = FREF * FCW \tag{4.1}$$



Figure 4.1: Simplified top-level diagram of low speed digital

Where the  $f_{outdiv}$  means the divided feedback variable high frequency signal. The number of bits of integer part of FCW determines the upper limit of the ADPLL's output frequency  $2 * f_{out_{div}}$  according to Eq.4.1, while its fractional part determines the resolution of the output frequency. The available reference frequency are 16 MHz and 32 MHz and targeted highest output frequency is at least 2.8 GHz. Thus 7 bits are allocated to the integer part. According to the modulation requirement, potentially 1 kHz should be available as the output resolution. Thus 16 bits are allocated to the fractional part of FCW.

The state machine is used to assist the bank switching between the three banks of DCO by determining the duration and order in which each of the three banks of DCO are activated after the frequency search is triggered. The block "tx interface" receives the tx data and the FCW from the digital baseband and generates the input control to the low-frequency and high frequency paths to facilitate two-point frequency modulation. The phase detector has four functions: (1) generate integer phase error by comparing output from both variable phase counter and reference phase accumulator. (2) Predict the next variable phase by accumulating DTC control words based on accumulated fractional reference phase. (3) Encode the Bang-Bang output since the Bang-Bang linear range is set to be double of DTC finest step size. (4) Observe the locking behaviour by looking at integer phase error and frequency error deviation. Based on (4) the loop can also be duty-cycled after lock-in at the cost of boosted in-band noise floor since the equivalent reference clock frequency is reduced. A separate loop filter for each of the three banks then processes the phase error to generate the appropriate control words for the DCO. Receiving the DTC control word, an LFSR based binary-to-thermal rotator encoder is implemented for relaxing the nonlinearity issue by DEM method. The implementation details of each of these blocks are described briefly in the following sections.

# 4.1.1. Digital Phase Error Detection Logic

The phase detector implemented in this ADPLL is shown in Figure 4.2. A difference mode architecture is adopted here: the frequency information is extracted from the phase to generate frequency error and then the immediate frequency deviation information is then accumulated to produce the phase error. In this way, the instantaneous frequency deviation can be observed and function like freezing or duty cycle the PLL could be implemented to save power in lock-in state. Difference of two successive samples of the counter gets subtracted from the integer part of



Figure 4.2: Integer path of differential mode based Phase error detector.

the FCW to generate the instantaneous integer part of the frequency deviation and when the loop finally locks in, the integer part of the phase error would be zero. For the fractional part, a combination of DTC and Bang-Bang detector is used. As discussed in Chapter 2, the accumulated value of the fractional part of the FCW generates the DTC control words. The overflow resulted from the accumulation of fractional part of FCW is added to the integer FCW. The scaling by " $inv_{kdtc}$ " is required to convert the normalization factor of the delay from divider-by-2 various period (ckvd2) to the DTC unit delay. This value of " $inv_{kdtc}$ " is obtained from the kdtc calibration block. The total phase error is obtained by adding the output of Bang-Bang phase detector to the integer part of the phase error. External control word is set up to help test the DTC part alone during test.

For the Bang-Bang's output, either +1 or -1, has to be scaled by a gain which is the ratio between a fine DTC step to a ckvd2 period as this is its linear detection range.

# 4.1.2. TX Interface for Two-point Frequency Modulation

The counter-based ADPLL has a natural wideband FM capability which can be realized as a two-point modulation scheme according to [1]. In Figure 4.3, one data path directly modulates a DCO, while another path is introduced to the reference path for compensation to avoid the phase error impacted by modulating data. According to Chapter 2, former path (DCO path) has a high-pass characteristics while the latter one (reference path) experience a low-pass filtering. Once both of the pats are combined an all-pass transfer function is realized and data rate for BLE as

1 Mbps could be ensured without restriction.

Implementation of the TX interface is shown in Figure 4.3. On input side a 10-bit modulation data "TX data" and frequency control word FCW can be seen as well as a DCO gain for normalizing the modulation data.

For compensating path at the reference path, TX data is added to FCW directly. 10-bit data is correspond to a frequency resolution finer than 1 kHz which is more than enough. Added by FCW, the compensated control word is fixed at reference frequency which is PVT free and no normalization is necessary. On the other side, normalization realized by being scaled by  $\frac{1}{k_{DCO}}$  is needed. In order to meet the corresponding SNR, resolution of DCO finer than tracking bank (50 kHz) may be asked and this is implemented by introducing a 5-bit  $\Sigma\Delta$  modulator to achieve equivalent sub-2 kHz resolution ( $\frac{50}{2^5}$ ).



Figure 4.3: TX interface block for two-point modulation

# **4.1.3.** 5-bit ΣΔ modulator

A 5-bit first order  $\Sigma\Delta$  modulator is used to reduce the quantization noise from the DCO. Figure 4.4 shows its block diagram. The dithering clock is chosen as either reference clock or higher rate of divide-by-16 by using the output from variable

counter. It is basically Carry-propagate adder (CPA) architecture based and the dithering rate is based on the requirement of the noise profile.



Figure 4.4: 1st order  $\Sigma\Delta$  modulator.

# **4.1.4.** Dynamic Element Matching

Linear, nonlinear and random mismatch could always be observed in integrated circuit due to linear gradients or even some geometry uncertainties introduced in fabrication. In order to relax the fractional spurs issue resulted from mismatch in DTC delay chain, dynamic element matching (DEM) is introduced. DEM is a dynamic process that reduces the effects of component mismatches in electronic circuits by rearranging dynamically the interconnections of mismatched components so that the time averages of the equivalent components at each of the component positions are equal or nearly equal. By appropriately varying the mismatched components' virtual positions, the effects of mismatched components can be reduced, eliminated, or frequency shifted.

Compared to the DTC designed in [2], the thermal code based control logic used in this design (will be covered in Section 4.2) makes every unit delay cell weights the same in terms of the final output signal. Thus DEM techniques could be applied easily to the control logic of the DTC. Linear feedback shift register (LFSR) based encoder-rotator is used here for simplicity. As shown in Figure 4.5, the encoder-rotator consists of a 4-bit Galois LFSR which generates a pseudo-random number as a rotation index. Once the rotation control logic is turned on, the binary

to thermal DTC control words encoder starts to rotate the thermal output according to the rotation index from LFSR. An example is shown in Figure 4.6, the rotation index is 3 and the encoded thermal control code is right shifted by 3 bits. The comparison between with and without rotation in terms of fractional spur level is shown in Figure 4.7.

DEM implemented here is to pseudo-randomly possible usage patterns of the DTC each reference period such that the error arising from unit delay element mismatches is scrambled from cycle to cycle. Thus, DEM increases DTC linearity at the expense of introducing more noise inband at the output spectrum of the ADPLL. However, the additional noise introduced by DEM is not a serious issue especially when the fractional spur requirement becomes more important according to certain standards.



Figure 4.5: 4-bit Galois LFSR.



Figure 4.6: Principal of LFSR based DEM.



Figure 4.7: Simulated phase noise showing DEM effect.



Figure 4.8: Simulated results showing reduced spurs.

The effect of DEM could be observed from Figure 4.7 and Figure 4.8. The INL of the example in this simulation is 3 LSB.

# 4.1.5. Loop Filter

Figure 4.9 shows the top-level view of digital loop filter. In terms of area and accuracy, DCO's capacitor array is implemented in a coarse-fine way by three segmented parts: PVT, Acquisition and Tracking bank as done in [1]. The former two are basically assisting the frequency acquisition by large frequency steps while the last one determines the frequency accuracy of the DCO as well as the PLL output signal. Thus different filters are used to generate corresponding control words to the three bands according to the input phase error bits (PHE). As already mentioned in previous chapter, type-I loop consists of only a proportional path which is good regarding large bandwidth, fast lock at the cost of less filtering effect (20 dB/decade out-of-band attenuation) and thus is used here for the PVT and Acquisition bank. With additional integral path, type-II loop could filter the DCO noise with attenuation of 40 dB/decade outside bandwidth at the cost of narrower bandwidth as well as slower locking speed. Besides, switchable IIR filter is also implemented in addition as a potential solution for the case when less noise is preferred. The control blocks implement the functionality to generate the zero-phase restart signals at mode switch-over and also to freeze the tuning word once the DCO mode is changed. For the control blocks offer three features: 1. Freeze the corresponding oscillation control words which is the input of related capacitor bank. This is done by the bank selection indicator word *bank*<sub>sel</sub> to set which operation mode is chosen at the moment. 2. Generate zero-phase restart signal. This is done by use an OR gate fed by the output control words to DCO and a preset signal which is "1" under a certain operation mode. However, whenever *bank*sel changes, it would trigger a D flip-flop to generate a negative pulse with the width as a reference clock. This negative pulse is used to clear the contents of the phase error accumulator as the output would be reset to 0. 3. **Open loop test plan.** The *bank*<sub>sel</sub> signal with 0 value would choose the control word from SPI externally instead of passing the filtered PHE corresponding results to DCO. Thus the DCO open loop performance could be tested.



Figure 4.9: Top-level view of digital loop filter.

# **4.2.** Mixed Signal Implementation

Situated at the feedback path, mixed signal blocks such as DTC, Bang-Bang phase detector as well as counter play the determining role of phase detector for the function of ADPLL as one of the three classic blocks in Figure 1.2. Via it, error in phase domain is digitized and sent into digital loop filter for processing. If one compares designing an ADPLL to riding a horse, then the phase detector could be analogous to the harness as it is adjusting the output signal (horse's movement) to track the input control words (rider's mind).

# 4.2.1. Coarse-Fine DTC Design

A digital-to-time converter (DTC) performs the opposite operation of a TDC: It has an n-bit digital input and produces a continuous-time digital signal where the positions of the edges are determined by the digital control signal. Briefly speaking, it is a digital controlled delay line. DTC is less frequently seen compared with TDC in ADPLL related literatures but is the most essential component in the phase detector of this design according to Figure 2.17. However, the delay line (both voltage controlled and digital controlled) topic is really common and could be found everywhere: from delay line based TDC [3] to delay-locked loop (DLL), from VCO (ring oscillator) to pulse width modulation (PWM) [4]. The simple but meaningful block functions as a cornerstone in a lot of systems.

Regarding a delay line based DTC design, both system level (control logic) and transistor level (unit cell) concepts are discussed as follows:

#### a. Architecture of DTC

The DTC's resolution, complexity as well as power are mostly determined at system level. The requirement for DTC in this design is to cover more than the required range (800 ps which is corresponding to 1.2 GHz as the divide-by-2 of the output) with as fine as possible step size while burning as low as possible power. As set by the goal proposed in Chapter 2, the DTC is required to cover more than 800 ps at a resolution at the level of 2 ps around (reference thermal noise level) with power lower than 40  $\mu W$ . What's more, tolerable linearity has to be ensured for a certain low worst fractional spur, according to the analysis done in Chapter 2.

However, none of the exiting popular DTC structures implemented in DPLL could meet such a need at low power. In [2], the resolution is as coarse as 15 ps and the control logic is not simple as well. The low power of the DTC is achieved at the cost of smallest size inverters with intolerable nonlinearity. What's more, the control logic determines that delay is always generated from a certain direction along the delay chain. This means at some certain channel, certain cells are working much more often than others. This is rather undesired because certain mismatch pattern would be visited periodically and thus introduces strong spurious tones at output. In [5], the resolution is good but the structure is kind of complex in terms of the huge numbers of MOS capacitor needed and the strong nonlinearity without look-up-table (LUT) based calibration. Besides, due to the shunt capacitor based delay unit, power consumed by that structure is really not low. In other publications, such as [6], the DTC resolution is not fine enough and the power of 200  $\mu$ W is also too high compared with the proposed target.

In this design, all the design considerations of the resolution, linearity and power with conventional TDC design are shifted to DTC part as covered in Chapter 2. Nevertheless the sub-gate resolution of DTC seems really hard to be implemented.

Though techniques such as passive delay line [7] and resistive interpolation [8] could help to offer such a fine resolution, still these methods are either too passive or analog intensive and thus they are not area economical or digitally compatible. Nevertheless, delay line based DTC should still be the solution. Delay line could be generally divided into two categories as: absolute delay and relative delay in terms of the timing relation between input and output. Absolute delay means

$$\phi_{out} = \phi_{in} + n * t_d \tag{4.2}$$

where the  $\phi$  denominates the phase of input and output signal. n is the digital delay control word while  $t_d$  is the resolution of the delay line. Single delay line or pseudo-differential delay line based TDC [3] and common multiplexer based DTC (shown in Figure 4.10) all belong to this category. Yet sometimes the relative delay between input and output phase is of more interest and relative delay based delay line's working principal is

$$\phi_{out} + nt_{dealy0} = \phi_{in} + nt_{delay1}$$

$$d = t_{delay1} - t_{delay0}$$
(4.3)

where the equivalent resolution d is realized by the relative difference between two different delay states of the same delay unit. To achieve this relative delay in TDC, the so called Vernier delay line consists of at least two delay lines is required. In this way the mismatch between two different delay line would create serious problem once going into pico second resolution. Besides, fine resolution realized by such a structure means that incredible long delay line would be unavoidable and this may result in large power consumption as well as large mismatch. In [9], a coarse-fine TDC is proposed. In terms of noise and spur this is really well designed. However, due to the large time amplifier array as well as the complex interface between coarse and fine TDC, the power is prohibiting high for the BLE application.



Figure 4.10: Delay line DTC based on a large multiplexer.

In this design, DTC is used as an alternative block to replace TDC for tracking

the fractional phase changing. Bang-Bang Detector is used for the error detection. The equivalent resolution of this structure is determined by DTC step size. Unlike TDC circuit, with DTC we can realized the relative Vernier delay thinking by a single delay line only. In this phase predication scheme, any fixed delay between reference edge and delayed reference edge could be considered as a initial phase offset which does not impact the loop locking function. In this way we can tolerate relative delay instead of an absolute delay. Thus with a single delay line, sub-gate resolution is already possible as shown in Figure 4.11. Assuming all the delay cells are with two status (when the control signal is low, the corresponding delay is  $t_0$  while when the control signal is high the corresponding delay is  $t_1$ ). When no delay is desired, there will be a fixed delay as  $nt_0$  while additional delay according to the control words of "1" would increase at the step of  $(t_1 - t_0)$ . Here the equivalent resolution of the DTC is only  $t_1 - t_0$  and could be infinitely fine as long as the mismatch is tolerable. However, in order to cover the required range at a low power regime, a single fine delay line consists of more than 400 ( $\frac{800}{2}$ ) stages is impossible. Thus coarse-fine is introduced here to reduce the power consumption. What necessary is to reduce required number of stages as many as possible in terms of accumulated jitter, mismatch and power. Finally 16 stages of coarse DTC with 60 ps resolution and 32 stages of fine DTC with 2 ps resolution are implemented in the tape out as an optimized option according to simulation. In this design, only the ratio between coarse stage and fine stage needs to be taken care of. Power hungry blocks such as time amplifier are avoided compared with its counterpart coarse-fine TDC implementation [9]. Besides, in terms of the ratio calibration, it could be implemented simply by the LMS algorithm as discussed in [10].



Figure 4.11: Simplified concept of Vernier DTC

#### b. Unit Cell of DTC

Cascaded inverter based delay line is preferred in this design in terms of low power and low complexity as well as small area. One important feature of inverter 4. ADPLL Implementation in Transistor-Level

is its propagation delay  $\tau$ .  $\tau$  is determined by the drive strength of the inverter, represented by the equivalent drive resistance  $R_{eq}$ , and the load capacitance seen at the output as  $C_L$ . Generally  $\tau$  could be approximated by

$$\tau = \ln(2)R_{eq}C_L \tag{4.4}$$

where ln(x) is the natural logarithm of x.  $C_L$  here consists of self load and input capacitance of any block driven by it as well as the capacitance of interconnect wires.  $R_{eq}$  depends on the size of the transistors in the inverter and is different for NMOS and PMOS. The  $R_{eq}$  of a transistor when used with full-swing signal is given by

$$R_{eq} = \frac{3}{4} \frac{V_{dd} (1 - 7\lambda V_{dd}/9)}{\beta((V_{dd} - V_T) V_{dsat} - V_{dsat}^2/2)}$$
(4.5)

where  $\beta = k \frac{W}{L}$  and W and L are transistor's width and length respectively. The parameters k,  $V_{dsat}$  and  $\lambda$  and the threshold voltage  $V_T$  are technology dependent constants. According to the analysis, we can see the  $R_{eq}$  is proportional to L/W and within a delay chain consists of the same unit cells,  $C_L$  is approximately proportional to WL. Thus the  $\tau$  is proportional to  $L^2$  and independent of W in first-order.

Based on this conclusion, we can generally divide delay unit cells into two main categories: current starved based and shunt capacitor based. The basic idea of shunt capacitor based is to change the  $C_L$  by shunting a equivalent switchable capacitor at the output of an inverter as shown in Figure 4.12. On the other hand, current starved based technique is to manipulate the (dis)charging current of the load capacitor. As shown in Figure 4.15, it is actually changing the  $R_{eq}$  by changing the W/L ratio. However, the DTC in ADPLL mainly consumes dynamic power which is given by [11]

$$P = C_L V_{dd}^2 f \tag{4.6}$$

where f is the gate toggling frequency. This equation implies that to generate a large delay by increasing load capacitance would increase the power consumption. Thus the current starve based delay cell is chosen for coarse stage. Because it is not wisdom to cover the whole ckvd2 range by enlarging  $C_L$  via the shunt capacitor unit cell in terms of power. Besides current starve based structure is also featured with wide tuning range. Regarding capacitor based delay cell, it is more proper for implementing fine stage in terms of linearity. Benefiting from the single line Vernier concept, this coarse-fine DTC's resolution is totally free from low supply voltage

impact and is really proper for low power low supply application. (Resolution is not depending on gate delay and thus independent from supply level.)

#### c. Mismatch for DTC Unit Cell Design

Nonidealities in the CMOS production process cause variations on the properties of the transistors. Usually, global and local variations are considered separately. Global variations affect all transistors on one chip in the same way, but may be different on another sample of the same chip. Thus it is also called inter-die variations. Its effect could be reduced by correct DTC gain estimation and is not of my interest here.

However, local variations (also known as intra-die variations) changes the properties of each transistor. Local variations are different for every transistor, which leads to mismatch between components that were intended to be identical. This is the main source for the nonlinearity discussed in Chapter 2 as a main source for fractional spurs.

Local transistor variations mainly lies in variations in the threshold voltage  $V_T$  and the current factor  $\beta$ . According to Pelgrom's law , both variations can be assumed to have a Gaussian distribution with zero mean and the standard deviations,

$$\sigma_{V_T} = \frac{A_{V_T}}{\sqrt{WL}} \tag{4.7}$$

and

$$\frac{\sigma_{\beta}}{\overset{\wedge}{\beta}} = \frac{A_{\beta}}{\sqrt{WL}} \tag{4.8}$$

where  $A_{V_T}$  and  $A_\beta$  are technology constants and  $\stackrel{\wedge}{\beta}$  is the nominal value here. Though this formula has deviation from real case especially in 40nm technology node, we can still use it for a qualitative estimation during design. According to 4.4, under the assumption of variations are small and uncorrelated, then  $\tau$  could be approximated by Gaussian distribution with a mean value equal to  $\stackrel{\wedge}{\tau}$  with standard deviation as

$$\sigma_{\tau} = \sqrt{\left(\frac{\partial \tau}{\partial V_T}\Big|_{\beta, V_T}^{\wedge} \sigma_{V_T}\right)^2 + \left(\frac{\partial \tau}{\partial \beta}\Big|_{\beta, V_T}^{\wedge} \sigma_{\beta}\right)^2}$$
(4.9)

Conclusion could be derived as

$$\frac{\sigma_{\tau}}{\frac{\Lambda}{\tau}} = \frac{A_{\tau}}{\sqrt{WL}} \tag{4.10}$$

where

$$A_{\tau} = \sqrt{\frac{A_{V_T}^2}{(V_{ad} - V_T^{\wedge} - V_{dsat}/2)^2} + A_{\beta}^2}$$
(4.11)

The derivation in detail is attached behind in appendix. In first order this could be considered as a Pelgrom's law form except the fact that  $V_{dsat}$  is proportional to L which is not a constant. However, as L decreases in modern technology, contribution from  $V_{dsat}$  becomes trivial and even negligible in 40 nm. The important conclusion from 4.10 is that W should be enlarged in order to reduce mismatch while L should be keep small since  $\sigma_{\tau}$  is proportional to  $L\sqrt{L}$ .

Consider a N-cell delay line and all cells are identical with  $\tau_{\mu}$  as the delay mean value while  $\sigma_{\mu}$  as the standard deviation. Then it is easily to draw the conclusion that the largest standard deviation occurs at the end of the delay line and is

$$\sigma_n = \sigma_\mu \sqrt{n} \tag{4.12}$$

where n is the number of the delay cells turned on and being N (all cells are turned on for delay) gives the largest error. According to the differential nonlinearity (DNL) definition, it is

$$DNL_n = \tau_n - \tau_\mu \tag{4.13}$$

The integral nonlinearity (INL) is defined as

$$INL_n = \sum_{i=1}^n [\tau_i] - n\tau_\mu$$
  
=  $\sum_{i=1}^n DNL_i$  (4.14)

Thus  $INL_n$  has zero mean and a standard deviation proportional to  $\sigma_{\mu}\sqrt{n}$ . The analysis done above is a really meaningful guideline for transistor sizing.

#### 4.2.2. Fine Bank of DTC

Unit cell is implemented as shown in Figure 4.12. 32 of such shunt capacitor based unit cell construct the fine bank of the DTC with a resolution around 2 ps in order to ensure Bang-Bang phased detector works in noise regime. According to 4.10, the L is kept the smallest while W is enlarged to relax mismatch. The layout is done in an S shape to reduce the linear gradient introduced mismatch during processing, as shown in Figure 4.21. The width is enlarged till we get a satisfying mismatch (with a worst fractional spur estimated to be lower than -40 dB) as shown in Figure 4.13.



Figure 4.12: Unit cell of fine DTC.



Figure 4.13: Monte Carlo simulation of fine cell.



Figure 4.14: Monte Carlo simulation of coarse cell.

# 4.2.3. Coarse Bank of DTC

As shown in Figure 4.15, 16 stages of current starved unit cell construct the coarse bank of the DTC with a resolution around 58 ps in order to cover the required ckvd2 range. In order to offer a large delay step at low power, two control signals are added in order to get large delay step size. One is introduced at the NMOS of the first stage for signal pulling down and the other one is the introduced at the starving PMOS of the second stage for pulling up, mainly impacting the rising edge since we do not care about the timing information carried by the falling edge of reference clock.

The layout is done in an S shape to reduce the linear gradient introduced mismatch during processing, as shown in Figure 4.21. The mismatch based on Monte-Carlo simulation is shown in the Figure 4.14.

In order to have a taste of the potential INL and DNL in real case, the mismatch extracted from Monte Carlo simulation is modelled in the delay line in Verilog and the transfer function of the coarse-fine DTC is shown in Figure 4.17, together with INL, DNL 4.18. However, the INL is based on best-fit straight line method.

In order to let the bandwidth behaving stable-mainly determined by reference jitter-the jitter accumulated from the delay chain has to be rather low and the



Figure 4.15: Unit cell of coarse DTC.



Figure 4.16: Simulated DNL in coarse band due to layout.



Figure 4.17: Transfer function of DTC including mismatch.



Figure 4.18: Linearity of DTC including mismatch.

simulated result is shown in Figure 4.19, the integrated rms jitter is 156 fs while the phase noise is also far below reference's level (-140 dBc/Hz).



Figure 4.19: (1)JEE of DTC delay chain.(2) Phase noise of DTC delay chain.

The power of the whole DTC is only 20  $\mu$ W which is rather low. The INL would be as large as 4 ps.

# **4.2.4.** Input Buffer of DTC

The reference clock is fed externally to the DTC via a buffer. Square wave reference clock is available in the lab so DC coupled buffer is used here for simplicity. The phase noise of the FREF clock should be kept as low as possible since it contributes to the output noise floor. The phase noise contribution from the input reference buffer is mainly the flicker noise of the very first stage, and thus large length of transistors should be utilized. The next stage ensures rail-to-rail output to drive a long line from the pad to DTC, where it is buffered again. The structure is basically a cascaded two stage buffer and the phase noise accumulated from input buffer together with DTC delay chain is simulated. Based on the simulation, the phase noise is kept lower than the reference's floor so still the bandwidth and Bang-Bang transfer function is dominated by reference noise, as shown in Figure 4.20.

# **4.2.5.** Summary of the proposed DTC

The total power consumption of the coarse-fine DTC is less than 20  $\mu$  W with a tolerable linearity. Layout is done in S shape instead of linear way to reduce impact from linear gradient mismatch. The layout is shown in Figure 4.21.



Figure 4.20: Input buffer phase noise.

In this design, the fractional phase error actually is tracked by a 2-stage DTC instead of complex TDC at a good resolution and linearity with low power. Bang-Bang Phase Detector is mainly situated for giving the feedback information. Compared with coarse-fine TDC proposed in [12], the DTC based phase detector is really enjoying the benefits from phase prediction.

# **4.2.6.** Supply Noise Monitor of DTC

However, one thing we need to consider is the impact from supply noise on this single delay line DTC. A dirty ground as well as supply noise would make the fixed delay of the single delay line based DTC bouncing together with the ground and thus modulating the input of the Bang-Bang phase detector. For low power purpose, a simple solution is to introduce a dummy delay line as a 1st order noise "monitor". As long as the dummy line experienced by  $CKV_{cly}$  is sharing the same ground as well as supply, the global bouncing effect is reduced to a large extent. This is shown in Figure 4.22

However, the high frequency could not be fed directly into the dummy delay line otherwise the power would be extremely high. Thus special care has to be taken



Figure 4.21: DTC top-level layout.



Figure 4.22: Supply noise monitor concept for single delay line based DTC.

here. The high frequency variable clock CKV could be first clock gated according to reference's edge, as shown in Figure 4.23 (CKV has to go through dummy path before time freezer).



Figure 4.23: Gated CKV generation.

Then gated CKV "ckv\_gated" is fed into the replica supply noise monitor as mentioned above, and shown in Figure 4.24. Now a supply noise monitor implemented by a clock gated replica delay path with low power is achieved in a low power way. Since only one OR gate is toggling at high CKV frequency, the power is really low with this structure (delay line is still toggling at reference clock rate). In a word, the potential supply noise introduced spur is suppressed for the first order.



Figure 4.24: Replica for noise cancellation.

# **4.2.7.** Bang-Bang Phase Detector

A single TSPC (shown in Figure 4.26) is preferred due to its compatibility with digital flow as well as considering the limited time for implementation. However, there are two variable periods before its output being sampled (depicted in Figure 4.27), thus plenty of time is ensured for the DFF to reach a well-defined output level. In this way metastability is not a serious issue in this design.

# 4.2.8. Counter Design



Figure 4.25: Synchronous counter basic cell [13].

Counter block performs the tracking of the integer phase of the variable clock by counting the number of ckvd2 clock cycles. As discussed previously, a 7-bit integer counter is required. The counter could be implemented either as synchronous or asynchronous logic. In a synchronous counter, all the flip-flops are clocked by the input clock. This may leads to a large power consumption but the outputs are synchronized and the delay is small. Though a asynchronous counter ( it behaves like a chain of divider-by-2) can save power, the robustness is not satisfying. It is sure that the output of each stage is only available after the preceding stage's output changing with some delay. In order to compensate such a delay, long delay chain is required for each digit output to ensure resampled clock can sample the right output. However, the delay chain is PVT sensitive and the delay chain for the first several high frequency bit would burn a lot of power. Once MSB is sampled at a improper time then the PLL may loss lock.

Thus from robustness aspect, a synchronous counter is implemented as shown in Figure 4.25. Though effort spent on making it contain standard cell only, the DFF in 40nm can not handle frequency of 1.5 GHz under 0.8 V in post-layout simulation. Thus custom designed TSPC DFF is used to replace the standard DFF. The TSPC

is shown in Figure 4.26. Though the core part burns more power than the asynchronous one, it still wins in terms of the fact that no delay compensation needed and the final power for the counter is 60  $\mu$  W.



Figure 4.26: Conventional TSPC DFF.

# **4.2.9.** Time Freezer Design

According to Eq. 4.6, the power consumption is proportional to the signal frequency. Thus the operation frequency of feedback phase detector needs to be clock gated since its output is only sampled at the reference rate. The snapshot clock gating trick is a main reason for reference design [2] to achieve sub-mW power. Thus this technique is reused also in the thesis design to clock gate down DCO output signal as one of the input of Bang-Bang phase detector. The generic thinking for the retiming logic is shown in Figure 4.27. The delayed reference clock rising edge "FREF<sub>dlv</sub>" would open a window transparent for the rising edge of CKVD2 to generate the frozen signal CKVD2f. Then after two CKVD2 clock, CKR is generated as the resampled reference clock. Time margin of two CKVD2 clock periods (1.6 ns) is ensured to avoid metastability, realized by DFF sampling. After these the window is closed for CKVD2 . This scheme is like the "bullet time" or "time freeze" in Hollywood movies which is characterised by the extreme transformation of time to be slow enough to show normally imperceptible events or movements. Here the low frequency CKVD2f generation is just a transition in time to detach the difference between FREF<sub>dly</sub> and high frequency frequency CKVD2, which is like to freeze phase error information in time domain.



Figure 4.27: Timing diagram of time freezer.

However, there is always an additional phase offset introduced during the freezing, which is to say the phase error between two input signal is generally enlarged due to different logic gates they are passing in this block. This would cause problem for phase error quantizer since once the distortion is so large that it is away from linear detectable range then the loop could not lock at all. In [2], delay chains of buffers are introduced for compensation. However this is really PVT sensitive which is still risky for the PLL's locking function. In this design, a dummy logic path as the one faced by *CKVD2* is added for *FREF*<sub>dly</sub> to generate a compensation signal as shown in the top path in Figure 4.28.

# 4.3. RF Implementation

DCO is the soul of ADPLL as it offers the most important output carrier signal. As shown in Figure 1.2, a DCO is at the heart of an all-digital phase-locked loop (ADPLL). With a digital input called as oscillator tuning word (OTW), a sinusoidal output with frequency proportional to the OTW could be gotten. Metrics such as power consumption, area, and phase noise profile are largely depending on this block. It is the horse when comparing designing of ADPLL to riding a horse, without whom there is no meaning to talk about "PLL" at all (actually it then becomes a delay-locked loop (DLL)).

General design consideration would be discussed first in this section, followed by implementation of key building blocks such as inductor, gm pair and capacitor



Figure 4.28: Time freezer.

array. Post-layout simulation results would be also shown to prove the design idea.

#### **4.3.1.** DCO Design

In terms of limited time and low power target, most of the effort for DCO design is spent to reducing the power based on conventional architecture for safety. Thus the classic capacitor array based LC DCO is realized here as depicted in [1]. Compared with DAC based LC DCO (DAC+VCO), this structure is more compatible with digital loop filer and more proper for advanced technology nodes such as 40 nm here. Negative resistance is used to compensate the loss from the parasitic in LC tank to sustain oscillation. The frequency tuning could be realized by changing the capacitance rather than the inductor for convenience in monolithic implementation. Switched Inductor would also face even lower quality factor (Q) issue. Hereby, the digital control is realized by a capacitor array tuned by OTW. The frequency of the DCO with N capacitor cells and an inductance L is given by

$$f = \frac{1}{2\pi \sqrt{L \bullet \Sigma_{i=0}^{N-1} C_i}}$$
(4.15)

In this work, a DCO covering a frequency range of 2.2 to 3 GHz with a phase noise of -110 dBc/Hz at 1 MHz offset from the oscillating frequency and a raw frequency resolution of 50 kHz is targeted. The design goal is to achieve this phase noise performance over the required tuning range with minimum power consumption. The design variables are the choice of inductor and capacitor values, width and length of the transistors.

#### a. Inductor

The power dissipated in the LC tank needs to be compensated so as to sustain the oscillation, which places a lower limit on the power consumption of the DCO. This loss power can be calculated by a simple tank model. The loss in the inductor due to the finite resistance of the metal is represented as a series resistance  $R_s$ . The quality factor of the on-chip inductors is usually worse than that of the capacitor banks and hence they dominate the overall energy loss of the LC tank. Hence, the capacitor losses are neglected. The negative resistance is used to model the active element that compensates the LC tank loss. The maximum energy stored in the inductor and capacitor is equal and is given by

$$E_{max} = \frac{1}{2}LI_{peak}^{2} = \frac{1}{2}CV_{peak}^{2}$$
(4.16)

where the  $I_{peak}$  and  $V_{peak}$  represent the peak values across the LC tank. The power loss due to  $R_s$  can be expressed as

$$P_{loss} = \frac{1}{2} I_{peak}^2 R_s$$

$$= \frac{1}{2} \frac{C V_{peak}^2 R_s}{L}$$

$$= \frac{1}{2} \frac{V_{peak}^2 R_s}{L^2 \omega^2}$$

$$= \frac{1}{2} \frac{V_{peak}^2}{L \omega Q_L}$$
(4.17)

Where  $\omega$  is the operating frequency and  $Q_L$  is the quality factor of the inductor defined as the ratio between imaginary part of the impedance over its real part.

$$Q_L = \frac{\omega L}{R} \tag{4.18}$$

According to Eq. 4.17, we can see that larger inductance and better Q is in favour of lower power consumption. However, regarding choosing the inductance, upper limit has to be set. Because for a certain center frequency, larger L means smaller headroom for capacitance to be tuned which limits the tuning range of the DCO. What's more, a larger L means more coupling capacitance to the substrate and thus a lower self-resonant frequency, beyond which the coils could not be considered as inductor anymore. Regarding Q, it is more or less limited by the technology. After estimating the amount of fixed parasitic capacitance from the capacitor banks and active part from post-layout extraction, a value of 8 nH is chosen after some iterations. Figure 4.29 shows the plot of the inductance as a function of frequency. The self-resonance frequency, beyond which an inductor behaves like a capacitor, is sufficiently far (>5.5 GHz) from the required operating frequencies (2.3–3 GHz).

Figure 4.30 shows the plot of the quality factor of the inductance with frequency. The quality factor is slightly less than 19.

#### b. Active Part

In terms of reducing power, it seems to be a golden rule that the NMOS-PMOS complementary pair would achieve lower power compared with single switch structure such as NMOS-only in term of the fact that the current can get reused by additional gm. However, this is the truth for high supply. Due to the additional PMOS pair, voltage headroom is reduced and can not win anymore under low supply







Figure 4.30: Q over frequency.

such as 0.4 V in GlobalFoundries 40 nm-LP. Besides, the output swing is also limited by the PMOS. However, for single switch such as NMOS-only structure, the swing could be as high as double of supply since the oscillator is DC biased at VDD. In this ADPLL, the dynamic divider can be used directly as a buffer for the feedback loop (In order to be connected to PA in TX path, additional AC coupled buffer is required for sure but it does not contribute power to the PLL budget anymore). However such a dynamic divider has a requirement for a larger input driving swing compared with CML based divider. Considering the buffer in [2] consumed more than 130  $\mu$ W finally, NMOS-only structure is adopted here for a peak-to-peak swing at least as large as 450 mV to drive the divider directly without a DCO buffer. The supply can be biased as low as 0.35 V. Thus finally much lower power is consumed in the RF part. The implemented circuit is shown in Figure 4.31. Current source is also removed for the lower DC bias.



Figure 4.31: NMOS only DCO designed.

#### c. Capacitor Array

Design of a capacitor array is rather challenging since a tuning range from 2.2 GHz to 2.8 GHz has to be covered with step size as fine as 50 kHz. This would result in an inconceivable huge array if thermal structure is adopted. Thus segmented structure is popular by dividing the capacitor array into 3 banks: PVT bank to cover

the full required tunable range with margin with most coarse step, Acquisition bank to cover one PVT step with much more margin at medium step and Tracking bank to cover a acquisition step with far more headroom at the required resolution. If even finer resolution is required then  $\Sigma\Delta$  modulator could be used.

The step size requirement could be related to the relationship as

$$\Delta f = (2\pi^2 L f^3) \Delta C \tag{4.19}$$

Based on Eq. 4.19, the resolution of each bank is determined by the resolution of the capacitor array.

Another important requirement of the capacitor banks is their quality factor. The total tank quality factor is given by the mathematical parallel combination of the inductor and capacitor quality factors. Hence, quality factor of the capacitor banks should be made larger compared to that of the inductor so that overall Q is limited by that of the on-chip inductance. As mentioned earlier, a Q around 20 for the inductor is achievable. Hence, a quality factor at least as large as 50 is targeted for the capacitor banks.

#### c.1 PVT Bank

The PVT bank together with the mode-selection bit determine the tuning range of the DCO. Shown in Figure 4.32, a switched MoM architecture is used here as the unit cell because it can provide large on-off step size.



Figure 4.32: PVT bank unit capacitor cell.

When the switch is on, the equivalent capacitance is a parallel of two  $C_{MOM}$  so we can get  $\frac{C_{MOM}}{2}$  as the on capacitance while when the switch is off, the capacitance over the two ends is determined by the parasitics of the switch and MoM capacitors.

In on state, Quality factor of the capacitor is degraded mainly by on resistance from the switch. Recall  $R_{eq}$  as discussed previously, we need to increase the aspect ratio or the overdrive voltage. To maximize the overdrive voltage, two pull-down NMOS transistors are used to pull the DC voltage of source and drain terminals of the MOS switch to ground in ON state. The aspect ratio of the MOS switch is then increased until its  $Q_{on}$  reaches 50.

Under off state, we have

$$C_{off} = \frac{C_{MOM}C_{parasitic}}{C_{MOM} + 2C_{parasitic}}$$
(4.20)

Where the  $C_{parasitic}$  is mainly contributed from drain and source junction capacitors of the NMOS switch, drain junction capacitance of the pull-down transistors, and the parasitics from the MoM capacitor to substrate. The Q in the off-state is largely determined by that of the MoM capacitors which is quite high.

The 5-bit control code is binary weighted and the number of unit cells controlled by n-bit is corresponding to  $2^n$ . The on-off capacitance is simulated based on extracted value from PVS.

Besides, quality factor is also ensured as shown in Appendix.



Figure 4.33: PVT bank layout.

Layout is shown in Figure 4.33. Capacitance, Q and cover range simulation results are attached in Appendix A.
#### c.2 Acquisition Bank

Structure same as PVT bank is used to implement the 6-bit medium resolution bank. The only difference is a smaller MoM capacitance is required now. The Q value can easily meet 50 target since  $Q_{on} = \frac{1}{\omega R_{on} C_{MOM}/2}$ .

The on-off capacitance is simulated based on extracted value from PVS.

Besides, quality factor is also ensure as shown below,

Layout is shown in Figure 4.34. Capacitance, Q and cover range simulation results are attached in Appendix A.



Figure 4.34: Acquisition bank layout.

#### c.3 Tracking Bank

Tracking bank is still MoM capacitor based due to its less vulnerability to PVT variations compared with MOS varactors.

As shown in Figure 4.35, under off-state the  $C_b$  is in series to  $C_s$  and the equivalent capacitance is

$$C_{off} = \frac{C_b C_s}{2(C_b + C_s)} \tag{4.21}$$

However when it comes to on state, the equivalent capacitance is  $\frac{C_s}{2}$  now and the capacitance step is then obtained by

$$\Delta C = \frac{C_s^2}{2(C_b + C_s)} \tag{4.22}$$



Figure 4.35: Tracking bank unit cell.

Table 4.1:	Capacitor	Array	/ Summary	/
------------	-----------	-------	-----------	---

Bank	Unit Cap	Number of bits
PVT	10 fF	5 bits
Acquisition	1.13 fF	6 bits
Tracking	14 aF	9(6MSB+3LSB)

From which we know that a smaller  $C_s$  and a larger  $C_b$  means the finer resolution. Tracking bank is segmented also in terms of less area and less fixed capacitance. The on-off capacitance is simulated based on extracted value from PVS.

Layout is shown in Figure 4.36. Capacitance, Q and cover range simulation results are attached in Appendix A.

The capacitor bank's performance could be summarised in Table 4.1.

#### d. DCO summary

The top-level DCO layout is shown in Figure 4.37. The size is 270  $\mu$ m \* 390  $\mu$ m which is mainly determined by the inductor.

The power consumption is only 168  $\mu$ W for the lowest output frequency while offering a peak-to-peak output swing larger than 500 mV, as shown in Figure 4.38.

From Figure 4.39 the phase noise is about -109 dBc/Hz at 1 MHz offset.



Figure 4.36: Tracking bank layout.

#### 4.3.2. Dynamic Divider

As shown in the system diagram before, a divider-by-2 is used right after DCO output to drive the phase detector block and make it easier for counter to operate at a lower frequency. Transmission gate based dynamic divider is chosen for its low power and low phase noise. Being able to be driven by the sinusoidal output from the DCO directly, this kind of divider can interact with phase detector directly without the buffer. Considering the big power consumption of the buffer in the loop, buffer is finally removed from the feedback loop and not included in the final power budget.

#### For Quarter Phase Assisted

The schematic of the dynamic divider used in this design is shown in Figure 4.40. The divider-by-2 contains four transmission gates and inverters connected in a loop. Each transmission gate is controlled by the differential input clock and only two of them are conducting in each half of the input clock period.

The most important reason to use this divider is not only because of the low power but due to the rail-to-rail output swing which can already interact with the following stages such as the time freezer. With the four phase, a DTC could also be done at the variable output path. This is depicted in Figure 4.41. By inserting a MUX, we can always choose the phase of the ckvd2 closest to the reference



Figure 4.37: DCO layout.



Figure 4.38: DCO transient simulation result.



Figure 4.39: DCO transient simulation result.



Figure 4.40: Transmission based dynamic divider.

edge. Instead of turning on more delay at the reference path, we can just choose the closest variable phase according to the accumulated fractional reference phase  $R_{R,f}$ 's relationship with 0.25, 0.5 or 0.75. Thus the required coarse DTC range is reduced.

However, the layout should be carefully drawn as shown in Figure 4.42.

In simulation and model, this architecture consumes similar power to the pure coarse-fine DTC structure and the mismatch from the four phase can bring serious fractional spur issue. Thus this method is not implemented in this tape out. The phase noise of the divider should be also kept much lower than the DCO's output as shown in Figure 4.43



Figure 4.41: Quadrature phase assisted Coarse-Fine DTC.



Figure 4.42: Layout of quadrature phase divider.



Figure 4.43: Simulated phase noise of divider.

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## 5

## ADPLL Top Level Completion, Simulation and Test Plan

In previous chapters, the ADPLL has been analysed in system level regarding basic transfer function and fractional spur estimation, with the insight into building blocks, simulation methodology and performance analysis. To forge all these design considerations into a feasible solution (finally a working chip) takes significant effort at the top level of the system. This chapter presents the top-level layout completion, summarized top-level simulation results and the final test plan.

#### 5.1. Top Level Layout

The top level layout with pin-out is shown in Figure 5.1. The chip size is 1.1 mm \* 1.1 mm. As can be seen in this figure, top-level the ADPLL is divided into three parts as explained in Chapter 4: DCO block (RF), DTC block (mixed signal) and Digital (digital logic and SPI interface). There are 5 different supplies:

**1.**  $VDD_{FRAC}$   $VDD_{FRAC}$  is the supply of the fractional phase error counter which is implemented as DTC and Bang-Bang phase detector here.

**2.**  $VDD_{RF}$   $VDD_{RF}$  is the low DC bias supply of the DCO.

**3.** *VDD*<sub>*DIG*</sub> *VDD*<sub>*DIG*</sub> is the supply for all the synthesized digital blocks.

**4.** *VDD*<sub>*INT*</sub> *VDD*<sub>*INT*</sub> is the supply of the variable counter to isolate its frequency toggling impact from supply sensitive DTC.

**5.**  $VDD_{BUF}$   $VDD_{BUF}$  is the supply for the output buffer since power contributed from output buffer is not included in this ADPLL design.

#### 5.2. Serial Peripheral Interface

The SPI block is an implementation of the slave device in the Serial Peripheral Interface (SPI) Bus with some registers. The SPI Bus is chosen as it is simple, flexible and can have a high data transmission rate (can work at MHz or tens of MHz). The SPI block is clocked by the external clock. Before the start-up of ADPLL, the SPI block will read the desired information from the off-chip controller via the signal line MOSI and then save the information to registers. The registers specify the select, enable and reset timing signals to the sequencer block or the low speed digital block or the blocks in the mixed signal part. Therefore the working mode and the status of ADPLL can be flexible. During the transient of ADPLL operation, some signals within the ADPLL system can be saved to the registers. The SPI block can read the saved signals and send them to the external controller via the signal line MISO, either for monitoring purpose or for test purpose. One extra input signal SPI RESET from pad is used to reset the registers. SPI is added on top of the digital core part and they are going through the digital flow together. Separate test bench is built to verify the write-in and read-out function of the synthesized SPI based on Verilog.

The SPI register table is available at A.

#### 5.3. Top Level Simulation Result

Top level simulation results are presented in this section regarding locking time, power consumption and phase noise.



Figure 5.1: Top-level ADPLL layout.

#### 5.3.1. locking behaviour

The locking behaviour is simulated both in Verilog based model and the AMS simulator based simulation as mentioned in Chapter 3. The frequency locking behaviour of channel FCW= 37.15 is shown in Figure 5.2. Besides, according to the sweeping of different channel in Verilog based simulation, the locking time varies from 20  $\mu$ s to 30 $\mu$ s from channel to channel.



Figure 5.2: ADPLL locking behaviour.

#### 5.3.2. Power

The power of digital part is simulated based on schematic level only. Other parts are simulated based on PVS extracted file with post-layout simulation. Th power budget is shown in Figure 5.3 as well as Table 5.1. The total power is less than 500 mW and from the pie diagram we can see this PLL's power consumption is mainly dominated by the DCO part.

#### 5.3.3. Output spectrum

According to theory analysis, this coarse-fine phase predictor based Bang-Bang ADPLL' in-band phase noise floor should be dominated by the levelled up reference noise. Besides, there should be no spur once the PLL is perfect with no nonlinearity

Block	Power (µW)
DCO+divider	210
DTC+BangBang	20
Low speed digital	110/165 (estimated as 1.5 times larger in real)
Counter+time freezer	65
Total	460 with digital 1.5 times enlarged for estimation

Table 5.1: Power results

as long as the gain is rightly estimated and this can be observed in Figure 5.4. Since the 2 ps resolution of DTC is at the same level as thermal noise, the potential spurs due to quantization are already randomized into noise. The black curve is the theoretically estimated result based on TDC based ADPLL model. Here, as mentioned before, the bandwidth of Bang-Bang based DPLL is shaped by the thermal noise and thus the real bandwidth is expected to be with slight difference compared with TDC based DPLL's theoretical bandwidth (depicted in black). The integrated RMS jitter is 600 fs now even in close integer channel. However, this could be explained: as long as the DTC is fine enough with good linearity, the Bang-Bang phase detector is almost working at an integer-N mode, and this is the channel which it is known for good performance.



Figure 5.3: ADPLL power consumption in pie chart.

Based on this, the PLL with no linearity issue would generate almost the same phase noise for each channel according to simulation.



Figure 5.4: Ideal ADPLL at FCW=38.001 channel.

The fractional spur level in the worst case with linearity issue according to Monte Carlo post-layout simulation is also good as shown in Figure 5.5. And the fractional spur's level of -45 dB could be explained by the formula derived in Chapter 2 with 2 dB deviation. This deviation is mainly resulted from the fact that the real INL is no longer ideal sinusoidal as assumed for the derivation. However, this level of spur is already good enough for most of the wireless standards (far beyond the BLE specification). The integrated RMS jitter is 1.2 ps around.

The fractional spur levels now in both close-integer channels and the rest channels are reduced to a satisfying level. Simulation is based on a reference input with thermal noise level at -138 dBc/Hz.

To sum up, for channel with worst fractional spur such as FCW=38.0001, the result is shown in Figure 5.5 with integrated RMS jitter as large as 1.2 ps. Other channels with fractional part larger than 0.001 are randomly checked and the integrated jitters are around 0.8 ps (other normal channels including the close integer one such as FCW=38.001 are more or less the same). Result for the case with channel FCW=38.001 is shown in Figure 5.6.



Figure 5.5: ADPLL at FCW=38.0001 channel as worst case.



Figure 5.6: ADPLL at FCW=38.001 channel as normal case.

#### 5.3.4. Comparison with state-of-art

There is no meaning to talk about FoM at this moment since everything is not measure yet. However, we can still calculate the FoM based on simulated results to see in the "best case", what this design would be in potential compared with others [1] [2] [3] [4] [5] [6] [7].



Figure 5.7: Comparison with state-of-art by FoM.

According to FOM definition as,

$$FoM = 10 \log[\frac{\sigma_{jitter}}{1s}^2 * (\frac{P}{1mW})]$$
(5.1)

In the worst case (1.2 ps RMS jitter together with 460  $\mu$ W power consumption), the estimated FoM could be as good as -241.7 dB while in the good case (0.8 ps integrated jitter) the FoM could be as good as -245 dB. However, certain deviation would always happen and that is why I put a red range around the best FoM point to represent a tolerable as well as normal deviation between simulation and measurement, considering noisy environment, complex nonlineairty and so on. This is shown in Figure 5.7.

#### 5.4. Test Plan

Assisted by SPI implementation, the separate block's test and internal signal observation are available:

**DCO open loop test** The open loop test mode of DCO is available as mentioned in Chapter 4.It is realized by the fact that the OTW can be read from the external DCO control signals via SPI from FPGA. This will be operated before the close-loop test of the ADPLL.

**DTC external control** The DTC's control word can be read from the external DTC control signals via SPI also from FPGA.

**phase error observation** The phase error generated from the digital phase error detection logic can be saved to the registers and read out also via the SPI.

**PCB** The PCB is implemented with OrCAD and the measurement would start from middle of December.

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# 

Conclusion

#### **6.1.** Conclusion for This Work

In this thesis, the analysis and implementation of a sub-half-mW all-digital phase locked loop (ADPLL) for frequency synthesis and modulation in ultra-low-power applications (BLE application) are presented.

In Chapter 2, it has been shown that TDC remains the main bottleneck in reducing the power consumption of ADPLL to sub-mW level. Besides, the fractional spurs issue is investigated. Basing on the conclusion drawn from the analysis, a novel coarse fine phase predictor based Bang-Bang DPLL is proposed to improve the phase detector's phase noise performance at low power. Thus fractional spur as well as power are greatly reduced in system level. Besides, DEM is available for further improving the linearity. Clock gating technique is realized by the time freezer block. Also, other tricks such as low power supply noise monitor as well as quadrature phase divider assisted DTC are also proposed and verified through simulation.

In Chapter 3, simulation methodology is presented, discussed and compared to the measurement results for verification.

In Chapter 4, circuit level implementation works are summarised. From the mixed signal aspect, sub-gate resolution DTC delay line is implemented in a simple but powerful way since the mismatch is reduced to a tolerable level with only 20 uW consumed. What's more, The RF part power is reduced by using a NMOS-only structure at low supply. With a large output swing at low power low supply, buffer is moved out from the PLL loop. Thus power is largely reduced. Regarding digital part, effort is spent in front-end (RTL to transistor netlist) flow for low power synthesizing. This also contributes to the final low power budget.

In a word, efforts spent to system level analysis, mixed signal design, RF design, digital flow, layout are paid back in terms of the reduced power, greatly enhanced fractional spur compared with the reference design as summarised in Chapter 5.

#### **6.2.** Future Work

The results of this thesis, leave plenty of future work to be done. Apart from actual measurement of the currently designed chips, a number of points require further investigation. These can be summarised as follows:

#### 6.2. Future Work

**Improving**  $K_{DTC}$  **estimation** Found during simulation and implementation, the gain of the DTC is extremely important for the locking and fractional spur performance. Due to limited time allowed, the least squired mean algorithm based calibration method is used. However a fast and accurate calibration would be really meaningful for this phase predication based ADPLL design. Because all the considerations as well as worries associated to the nonlinearity of the DTC part would be relaxed. However, in terms of this coarse-fine DTC, the ratio between coarse stage and fine stage is a part of the gain for the whole DTC block. Thus the algorithm still fits well with my structure. Anyway, further effort should be spent on this block to take fully advantages of the digital calibration friendly characteristic of DTC instead of conventional TDC.

**Improving DCO structure** As could be seen in Chapter 4, the class of this designed DCO is the conventional type with no special tricks for phase noise improvement. However, as another main contributor for not only the integrated jitter but also power, DCO is the soul of the ADPLL and further effort should be spent in DCO structure. For example, since we are targeting for low supply low power application, the transformer class-F DCO could be investigated to further push the limit of this proposed system.

# A

### Appendix

A.1. Capacitor Bank simulation



Figure A.1: PVT bank unit capacitor cell Capacitance.



Figure A.2: PVT bank unit capacitor cell Q.



Figure A.3: PVT bank swept capacitance.



Figure A.4: Acquisition bank unit capacitor cell Capacitance.



Figure A.5: Acquisition bank unit capacitor cell Q.



Figure A.6: Acquisition bank swept capacitance.



Figure A.7: Tracking bank LSB capacitor cell Capacitance.



Figure A.8: Tracking bank LSB unit capacitor cell Q.



Figure A.9: Tracking bank swept capacitance.



Figure A.10: Tracking bank MSB unit capacitor cell Capacitance.



Figure A.11: Tracking bank MSB unit capacitor cell Q.

### A.2. SPI register table

Register	BIT	Name	description	Default value
00	0	onon	turn on of divider strucutre	1 (off)
	1	spitdtc		0
	2	dcopath[0]		0
	3	dcopath[1]		0
	4	dcopath[2]		0
	5	bank_en[0]		1
	6	bank_en[1]		1
	7	bank_en[2]		1
01	0	inv_ktdc[0]		0
	1	inv_ktdc[1]		1
	2	inv_ktdc[2]		1
	3	inv_ktdc[3]		0
	4	inv_ktdc[4]		0
	5	inv_ktdc[5]		1
	6	inv_ktdc[6]		0
	7	inv_ktdc[7]		1
02	0	inv_ktdc[8]		1
	1	inv_ktdc[9]		1
	2	inv_ktdc[10]		1

	3	inv_ktdc[11]		0
	4	inv_ktdc[12]		0
	5	inv_ktdc[13]		1
	6	inv_ktdc[14]		0
	7	inv_ktdc[15]		0
03	0	DTC_coarse[0]	SPI	0
	1	DTC_coarse[1]		0
	2	DTC_coarse[2]		0
	3	DTC_coarse[3]		0
	4	DTC_fine[0]	SPI	0
	5	DTC_finec[1]		0
	6	DTC_fine[2]		0
	7	DTC_fine[3]		0
04	0	FCW[0]	DEFAULT 38 CLOSE IN	0
	1	FCW[1]		1
	2	FCW[2]		1
	3	FCW[3]		0
	4	FCW[4]		0
	5	FCW[5]		0
	6	FCW[6]		0
_	7	FCW[7]		0
05	0	FCW[8]		0
	1	FCW[9]		0
	2	FCW[10]		0
	3	FCW[11]		0
	4	FCW[12]		0
	5	FCW[13]		0
	6	FCW[14]		0
	7	FCW[15]		0
06	0	FCW[16]		0
	1	FCW[17]		0
	2	FCW[18]		1
	3	FCW[19]		0
	4	FCW[20]		0
	5	FCW[21]		1

	6	FCW[22]	0
	7		
07	0	inv_kdcomod[0]	1
	1	inv_kdcomod[1]	0
	2	inv_kdcomod[2]	0
	3	inv_kdcomod[3]	0
	4	inv_kdcomod[4]	1
	5	inv_kdcomod[5]	0
	6	mod_on	0
	7		
08	0	memdcop[0]	0
	1	memdcop[1]	0
	2	memdcop[2]	1
	3	memdcop[3]	0
	4	memdcop[4]	0
	5		
	6		
	7		
09	0	memdcoa[0]	0
	1	memdcoa[1]	1
	2	memdcoa[2]]	1
	3	memdcoa[3]	0
	4	memdcoa[4]	0
	5	memdcoa[5]	0
	6		
	7		
10	0	memdcot[0]	0
	1	memdcot[1]	0
	2	memdcot[2]	0
	3	memdcot[3]	1
	4	memdcot[4]	1
	5	memdcot[5]	0
	6	memdcot[6]	0
	7	memdcot[7]	
11	0	memdcot[8]	0

	1	1	
	1		
	2	alphaad[0]	0
	3	alphaad[1]	1
	4	alphaa[0]	0
	5	alphaa[1]	1
	6	alphap[0]	0
	7	alphap[1]	1
12	0		
	1	rho[1]	1
	2	rho[2]	1
	3	rho[3]	1
	4	rho[4]	0
	5	alphat[0]	0
	6	alphat[1]	0
	7	alphat[2]	1
13	0	pvt_mode[0]	1
	1	pvt_mode[1]	0
	2	pvt_mode[2]	0
	3	BUFF_en	1
	4	iiren	0
	5	lambda[0]	0
	6	lambda[1]	1
	7	lambda[2]	0
14	0		
	1		
	2	kdcop[0]	0
	3	kdcop[1]	1
	4	kdcop[2]	1
	5	abmode[0]	0
	6	abmode[1]	1
	7	abmode[2]	0
15	0	dtc_mu[0]	1
	1	dtc_mu[1]	1
	2	dtc_mu[2]	0
	3	dtc_mu[3]	1

A.3. Deriv	ation	for Mismatch for	DTC unit cell design	127
	4	dtccal		0
	5	kdcoa[0]		0
	6	kdcoa[1]		0
	7	kdcoa[2]		1
16	0	kdcot[0]		0
	1	kdcot[1]		1
	2	kdcot[2]		0
	3	kdcot[3]		0
	4	kdcot[4]		0
	5	kdcot[5]		1
	6	kdcot[6]		1
	7	kdcot[7]		0
17	0	Kres[0]		0
	1	Kres[1]		1
	2	Kres[2]		0
	3	Kres[3]		0
	4	Kres[4]		0
	5	Kres[5]		1
	6	Krest[6]		1
	7	Kres[7]		0
18	0	Kres[8]		0
	1	rotate_en		1
	2			
	3			
	4			
	5			
	6			
	7			

### A.3. Derivation for Mismatch for DTC unit cell design

Here is the details for the derivation procedure from Eq. 4.9 to Eq. 4.10. Bring Eq. 4.4 into Eq. 4.9, we have

$$\sigma_{\tau} = ln(2)C_L \sqrt{\left(\frac{\partial R_{eq}}{\partial V_T}\Big|_{\stackrel{\wedge}{\beta}, V_T}^{\quad \beta} \sigma_{V_T}\right)^2 + \left(\frac{\partial R_{eq}}{\partial \beta}\Big|_{\stackrel{\wedge}{\beta}, V_T}^{\quad \beta} \sigma_{\beta}\right)^2}$$
(A.1)

$$\frac{\sigma_{\tau}}{\frac{\Lambda}{\tau}} = \frac{1}{R_{eq}} \sqrt{\left(\frac{\partial R_{eq}}{\partial V_T}\Big|_{\beta, V_T}^{\Lambda, \Lambda} \sigma_{V_T}\right)^2 + \left(\frac{\partial R_{eq}}{\partial \beta}\Big|_{\beta, V_T}^{\Lambda, \Lambda} \sigma_{\beta}\right)^2}$$
(A.2)

Where 
$$\frac{\partial R_{eq}}{\partial V_T}|_{\beta, V_T}^{\wedge \wedge}$$
 is

$$\frac{\partial R_{eq}}{\partial V_T}\Big|_{\stackrel{\wedge}{\beta}, \stackrel{\wedge}{V_T}} = \frac{\stackrel{\wedge}{R_{eq}}}{V_{dd} - \stackrel{\wedge}{V_T} - V_{dsat}/2}$$
and
$$\frac{\partial R_{eq}}{\partial \beta}\Big|_{\stackrel{\wedge}{\beta}, \stackrel{\wedge}{V_T}}$$
is
$$(A.3)$$

$$\frac{\partial R_{eq}}{\partial \beta}|_{\beta, V_T}^{\wedge} = -\frac{R_{eq}}{\beta}$$
(A.4)

Submitting them into Eq. A.2 we can get the Eq. 4.10.