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2006

Link to publication

Citation for published version (APA): Dasalukunte, D., Palsson, A., Kamuf, M., Persson, P., Veljanovski, R., & Öwall, V. (2006). *Architectural Optimization for Low power in a Reconfigurable UMTS filter.* Paper presented at International Symposium on Wireless Personal Multimedia Communications (WPMC), 2006, San Diego, United States.

Total number of authors: 6

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Architectural Optimization for Low Power in a Reconfigurable UMTS Filter

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Abstract— This paper presents an improved architecture for an UMTS filter which reduces the switching power within the filter by 75% compared to the original design. The filter length is dynamically varied depending on the adjacent channel noise. This attains the least power consumption in noise free environments and required filter performance in the presence of noise. As a consequence, the battery life of the mobile device is improved without compromising with the 3GPP standard filter specifications. Furthermore, the design is simplified by reducing the number of clock domains from 3 to 2. In the improved design most blocks run at a slower clock, reducing switching power, and thus the overall power consumption.

Index Terms-UMTS, 3GPP, reconfigurable, low power

I. INTRODUCTION

W-CDMA (UMTS) is the third generation (3G) standard for wireless communication, which offers a variety of wideband data and multimedia services. The emergence of W-CDMA standard has increased the amount of computing on the mobile devices. These systems are required to consume minimum power in order to extend the battery life of the device. Furthermore, Adjacent Channel Interference (ACI) has become a critical issue due to the increase in information transmission within a finite bandwidth. Thus, design of digital filters in mobile terminal receivers has become a challenge with tight and complex filter specifications for minimal power consumption.

The receiver block diagram is shown in Fig. 1. The digital channel filter is placed after the RF front-end and analog-to-digital converter (ADC) and performs the required filtering in order to obtain the desired frequency spectrum. The resulting frequency spectrum is descrambled, despread, and demodulated to retrieve the data that corresponds to the mobile terminal user. This data is termed as desired signal. The UTRA-TDD, one of the two duplex modes developed by the 3rd Generation Partnership Project (3GPP) [1], has a chip rate of 3.84 Mega chips per sec (Mcps) and a frame has a duration of 10ms, with each frame consisting of 15 timeslots (2560 chip/slot) as illustrated in Fig. 2.

This paper starts by describing the original reconfigurable filter and the various processing blocks [2]. The following part focus on the work carried out to optimize the design for low power. Primarily, optimizations have been done at the architectural level.



Fig. 1. Block diagram of the mobile receiver terminal



2560 samples/timeslot

Fig. 2. Frame structure of the UTRA-TDD signal

II. THE RECONFIGURABLE FILTER

To achieve the performance of an UTRA-TDD receiver the digital channel filter length is required to be 65 taps [2]. However, in noise free environments such a high order filter is power inefficient. In [2] a reconfigurable filter was proposed and was implemented in [3]. The filter reduces power consumption in order to improve the battery performance of the mobile device. This is done by measuring the inband and out-of-band signal powers and using a control unit to calculate the required filter length. In case of noise free environments the out-of-band signal power would be low and hence a shorter filter length would be sufficient to filter out the noise, thus lowering power consumption. The reconfigurable filter in Fig. 3, consists of an FIR filter, three Signal Power Measurement (SPM) units and a control unit to vary the filter length depending on the out-of-band noise. The transmit filter employs an interpolation factor of 4 to satisfy the tradeoff between analog reconstruction filter and the complexity for the digital filter. This results in transmitting the data at 4×3.84 MHz = 15.36MHz. Therefore, the receiver filter has an input data rate of 15.36MHz followed by a decimation stage to downsample the data by 4.

A. The original FIR Filter

The FIR filter employs a maximum filter length of 65 and a minimum length of 5. This has been derived in [2] in order to satisfy the filter specifications of the UTRA-TDD receiver



Fig. 3. The Reconfigurable filter



Fig. 4. The 3GPP filter specifications

shown in Fig. 4. A minimum stopband attenuation of 33dB is to be maintained according to the 3GPP specifications. Symmetric coefficients have been exploited to fold the filter and minimize multiplications [4]. Further optimizations on the filter coefficients have been done in [3] by adopting a bit-shift-fitting algorithm to convert multiplications with coefficients into bit-shifts. The filter also employs logic to switch the taps ON/OFF as directed by the control unit.

B. The Signal Power Measurement Unit

The Signal Power Measurement (SPM) unit consists of a full wave rectifier (FWR) and a first order infinite impulse response (IIR) filter. The block diagram of the SPM unit proposed in [2] and implemented in [3] is shown in Fig. 5. The IIR filter accumulates the absolute value of the signal from the FWR over a period of time to calculate the signal power. A truncation unit T is used just before the output is fed back into the system to maintain fixed wordlength [5]. The SPM unit calculates the signal power in every timeslot before starting afresh in the next timeslot. The filter coefficients δ



Fig. 5. The signal power measurement unit

and γ are calculated according to

$$\delta = \frac{\cos\theta}{1+\sin\theta} \tag{1}$$

and
$$\gamma = \frac{1-\delta}{2}$$
, (2)

where θ is the normalised frequency of 0.002π [2].

C. Control Unit

The control unit is the intelligence behind the reconfigurability of the system. It uses the inband, out-of-band and desired signal powers from SPM units to calculate the filter length. The desired signal is obtained by despreading the inband signal using the user's orthogonal variable spreading code. The Adjacent Channel Performance (ACP) derived in [2] from the E_b/N_0 model is calculated according to the equation:

$$ACP = \frac{P_{\text{outband}}}{P_{\text{desired}}(1 + \frac{P_g}{E_b/N_0}) - P_{\text{inband}} - \eta}$$
(3)

where P_{outband} , P_{inband} , and P_{desired} are out-of-band, in-band and desired signal powers respectively. The processing gain P_g and thermal noise η are constants, while the target E_b/N_0 is set. Using ACP and Adjacent Channel Leakage Ratio (ACLR) specified by the 3GPP, Adjacent Channel Selectivity (ACS) is calculated as:

$$ACS = \frac{1}{\left(\frac{1}{ACP}\right) - \left(\frac{1}{ACLR}\right)}.$$
(4)

The control unit adapts a lookup table based approach by using the new ACS value in order to determine the length of the filter. The control unit monitors the out-of-band and inband signal powers every timeslot and accordingly varies the filter length in order to keep the filter specifications such as stopband attenuation, consistent. The filter is reconfigured by switching the taps on/off depending on whether the filter length needs to be increased or decreased. The number of filter taps that need to be switched on/off is obtained from a lookup table for a particular value of ACS. Increasing the filter length is done at once, while switching off the filter taps is gradual. This is done to avoid poor filtering of the out-of-band signals when many taps are switched off and there is a surge in noise levels. This is termed as hysteresis protection [2].



Fig. 6. The Optimized Filter

III. OPTIMIZED ARCHITECTURE

The FIR filter occupies more than 50% of the entire design and is always active. By moving the decimators prior to the filter, the filter can be run at a lower clock, resulting in power savings due to reduced switching activity.

Since the decimation factor is 4, the FIR filter would be divided into four filterbanks [5], each $1/4^{\text{th}}$ the length of the original filter. The optimized architecture of the reconfigurable filter is shown in Fig. 6 in which the decimators now appear at the input and the filterbanks are denoted FIR1, FIR3, and FIR24. By doing so, the arithmetic operations are now reduced to a fourth. The coefficients corresponding to these banks are shown in Fig. 7. The coefficients corresponding to the 1^{st} and 3^{rd} filterbanks have inherent symmetry, while 2^{nd} and the 4^{th} filterbanks are merged together to achieve coefficient symmetry. This halves the number of multiplications since the filter can be folded.

A. Clock domain reduction

Under the assumption that integration in the SPMs had to be performed over an entire timeslot, the previous implementation [3] of the reconfigurable filter utilized 3 clocks. Two clocks, 15.36MHz and 3.84MHz are required because of different input and output data rates. Calculations within the control unit involve divison operations to calculate ACP as in (3), and the new filter length. As a result, a faster clock was needed to complete the new filter length calculations before the data from the next TIMESLOT arrived. Thus, the third clock was needed for the control unit. The frequency was estimated to be 32 times 3.84MHz, i.e., 122.88MHz, as the control unit required 32 clock cycles to perform division and other operations to obtain the ACS [6].

However, through simulations it has been found that the SPM units saturate quite early, see Fig. 8, which shows the signal power for data in one timeslot. The early saturation of the SPM units is used to initiate the control unit to start with the new shaver value calculations a few clock cycles before the last data sample arrives. As the later samples do not contribute to the measurement of signal powers significantly the calculation of new filter length by neglecting the last few data samples introduces almost no errors. This results in the



Fig. 7. Coefficients of the filterbanks equivalent to the original filter coefficients

control unit to run on a slower clock, i.e., at 15.36MHz instead of 122.88MHz. The reduction in number of clock domains is important as it makes hardware implementation easier. The reduced clock frequency also lowers the demands on the hardware blocks being implemented. The control unit clock can further be reduced and run at 3.84MHz by neglecting more number of data samples in the received TIMESLOTs. However, the number of clock domains cannot be reduced further because of different input and output data rates of the system.

B. SPM Optimization

The SPM units are implemented in the same way as in the original design, but with a few more optimizations. The coefficients obtained from (1) and (2) are $\delta = 0.9937$ and $\gamma = 0.0031$, respectively. Scaling the coefficients by 2^8 , instead of 2^{12} as in the original design, results in $\gamma \approx 1$, and one multiplication can be omitted. Thus, one out of two coefficient multiplications are avoided in each of the three SPM units. However, the performance or the functionality as compared to the original SPM unit is unaffected with this optmization.



Fig. 8. Early saturation of SPM unit

 TABLE I

 Power estimation results from the two designs

	Switching	Internal	Total	
	power	power	power	
Optimized (3.84MHz)	0.0368 mW	0.069 mW	0.106 mW	
Original (15.36MHz)	0.15 mW	0.281 mW	0.431 mW	

IV. IMPLEMENTATION AND RESULTS

The complete system was initially modeled and simulated in MATLAB. The improved architecture was designed and implemented in a 0.35μ m standard CMOS process. The photo of the fabricated chip is shown in Fig. 9 and it measures 2.7mm×2.1mm. In the optimized design, the FIR filter has contributed to a significant reduction in power, due to the moving of decimators and hence running it 4 times slower. The original design was implemented in an FPGA but simulated for power estimation using NEC $0.25\mu m$ process [3] [6]. To compare the power consumption of both architectures 130nm CMOS standard cell libraries were used. The FIR filter occupied a large portion of the entire design and it was the one that was optimized significantly. So the comparison in power consumption was done by comparing the two FIR filters, one running at 15.36MHz and the other at 3.84MHz and the results are presented in Table I. By comparing the various parameters between the designs, Table II, it can be observed that the FIR filter now runs at a lower clock frequency. This could be further exploited to use a lower power supply. The control unit clock has also been reduced, in turn reducing the highest clock frequency from 122.88MHz to 15.36MHz. The reduction in clock domains contribute significantly during hardware design as it is simpler and easier to manage designs with fewer clock domains.

TABLE II Comparison between the two designs

Parameter	Original Design	Improved Design
No. of clock domains	3	2
Control unit clock	122.8MHz	15.36MHz
Highest Clk frequency required	122.88MHz	15.36MHz
FIR filter clock	15.36MHz	3.84MHz
No. of decimators	2	4
No. of multipliers in SPM unit	2	1



Fig. 9. Chip microphotograph of Reconfigurable UMTS filter

V. CONCLUSION

As it is difficult to do a fair comparison on the power figures obtained from two different processes on which the original $(0.25\mu\text{m})$ and the optimized $(0.35\mu\text{m})$ architectures were implemented. The power consumption in the two architectures has been done by choosing a common process (130nm). The dynamic power consumption reduced by a factor of 4, by moving the decimators prior to the filter. The control unit clock was also reduced by exploiting the early saturation of the SPM units which in turn reduced the number of clock domains. Optimizations performed on the SPM units by a better representation of the coefficients reduced the multiplications from 2 to 1. The overall reduction in power is due to the reduced switching activity in the FIR filter.

REFERENCES

- "Technical Specification Group Radio Access Networks Physical Layer -General Description 3G TS 25.201 version 3.0.2," 3rd Generation Partner Project, Tech. Rep., 2000.
- [2] R. Veljanovski, "A Reconfigurable Root Raised Cosine Filter for a mobile receiver," Ph.D. dissertation, Victoria University of Technology, 2003.
- [3] H. Bruce, "Power optimisation of a reconfigurable FIR filter," Master's thesis, Lund University, Sweden, 2004.
- [4] K. K. Parhi, VLSI Digital Signal Processing Systems. New York, NY: Wiley, 1999.
- [5] S. K. Mitra, Digital Signal Processing: A Computer-Based Approach, 2nd ed. McGraw-Hill, 2001.
- [6] R. Veljanovski, J. Singh, M. Faulkner, and V. Owall, "Design and ASIC performance analysis of a reconfigurbale digital filter for UMTS," in *Proc. Seventh International Symposium on Signal Processing and its Applications*, vol. 2, Jul 2003, pp. 55–59.