

Optimal Multi-objective Approach for VLSI Implementation of Digital FIR Filters

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Abstract— Filters are heart of any Digital Signal Processing (DSP) based system wherein, multipliers and adders are the basic component in Finite Impulse Response (FIR) filters. So, VLSI implementation of DSP system's performance is generally determined by the performance of the multipliers and adders. Multiplier is generally the slowest element in the system. Furthermore, it is generally the most area consuming. But optimizing the speed and area of the multiplier is a major design issue because improving speed results mostly in larger areas.

So, in this paper, an optimal multi-objective approach for VLSI implementation of digital FIR filters is suggested, wherein the three main design constraints viz. area, speed and power are optimized simultaneously without affecting the functionality of design.

Keywords— FIR filter, direct form FIR filter, transpose form FIR filter, MAC, Multiplier, MCM, SCM.

I. INTRODUCTION

Finite impulse response (FIR) filters are of great importance in digital signal processing (DSP) systems since their characteristics in linear-phase and feed-forward implementations make them very useful for building stable high-performance filters.

Physically, a discrete system (FIR filters) is realized or implemented either as a digital hardware or as software on a digital hardware. The processing of the discrete time signal by the digital hardware involves mathematical operations like addition, multiplication and delay.

In signal processing, a finite impulse response (FIR) filter is a filter whose impulse response (or response to any finite length input) is of finite duration, because it settles to zero in finite time. This is in contrast to infinite impulse response (IIR) filters, which may have internal feedback and may continue to respond indefinitely (usually decaying). [1]

The time domain representation of N^{th} order FIR system is,

$$h[n] = \sum_{i=0}^N b_i x[n - i] \quad (1.1)$$

The impulse response $h[n]$ of digital FIR filter can be calculated if we set $x[n] = \delta[n]$ in the above relation, where $\delta[n]$ is the Kronecker delta impulse. The impulse response for an FIR filter then becomes the set of coefficients b_n , as follows:

$$h[n] = \sum_{i=0}^N b_i \delta[n - i] = b_n; \text{ For } n=0 \text{ to } N \quad (1.2)$$

The Z-transform of the impulse response yields the transfer function of the FIR filter i.e.

$$H(z) = Z\{h[n]\} = \sum_{n=-\infty}^{\infty} h[n]z^{-n} = \sum_{n=0}^{N-1} b_n z^{-n}; \quad (1.3)$$

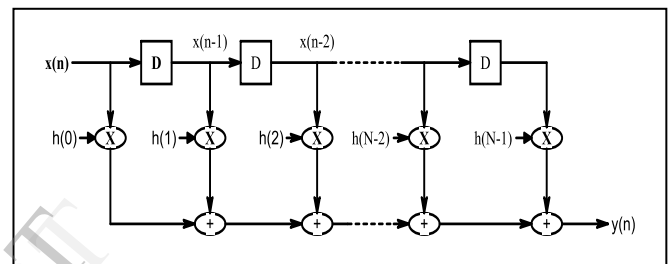


Fig. 1.1: A discrete-time FIR filter of order N. The top part is an N-stage delay line with $N + 1$ taps

FIR filters are clearly bounded-input bounded-output (BIBO) stable, since the output is a sum of a finite number of finite multiples of the input values, so can be no greater than $\sum |b_i|$ times the largest value appearing in the input [2].

II. WHY FIR FILTERS

An FIR filter has a number of useful properties which sometimes make it preferable to an infinite impulse response (IIR) filter. FIR filters:

- i. Require no feedback. This means that any rounding errors are not compounded by summed iterations. The same relative error occurs in each calculation. This also makes implementation simpler.
- ii. Are inherently stable. This is due to the fact that, because there is no required feedback, all the poles are located at the origin and thus are located within the unit circle (the required condition for stability in a discrete, linear-time invariant system).
- iii. They can easily be designed to be linear phase by making the coefficient sequence symmetric; linear phase, or phase change proportional to frequency, corresponds to equal delay at all frequencies. This property is sometimes desired for phase-sensitive applications, for example data communications, crossover filters, and mastering.

III. BASIC BUILDING BLOCKS OF FIR FILTERS

A. MAC Unit

From figure 1, it is seen that the critical operations usually involve are many multiplications and/or accumulations. Hence for real-time signal processing, a high speed and high throughput Multiplier-Accumulator (MAC) is always a key to achieve a high performance digital signal processing system.

A conventional MAC unit consists of multiplier and an accumulator that contains the sum of the previous consecutive products. The function of the MAC unit is given by the following equation:

$$F = \sum A_i B_i \quad (3.1)$$

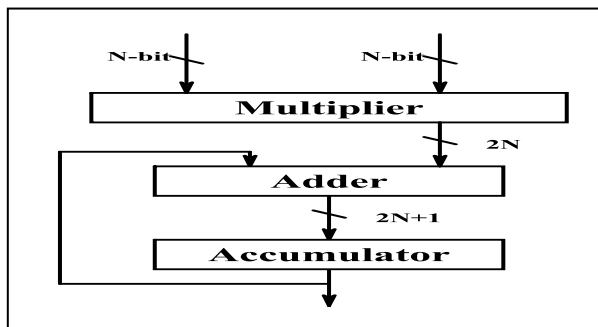


Fig. 3.1.: Basic structure of MAC

B. ADDER UNIT

In electronics, an adder is a digital circuit that performs addition of numbers. In modern computers adders reside in the arithmetic logic unit (ALU) where other operations are performed. Depending on the area, delay and power consumption requirements, several adder implementations have been proposed. Ripple Carry Adders with the most compact design ($O(n)$ area) among all types of adders, are the slowest in speed ($O(n)$ time). Carry Select Adders ($O(n)$ time) and ($O(2n)$ area) are in between RCAs and CLAs ($O(n)$ time) and ($O(n \log n)$ area) thus providing an optimum solution between the area-efficient RCAs and the shortest-delay CLAs.

IV. DESIGN ISSUES

The main goal of a DSP processor design is to enhance the speed of the MAC unit, and at the same time limit the power consumption and number of gates (or area).

There are three sources of power dissipation in CMOS circuits: switching power P_{sw} , short-circuit power P_{sc} , and leakage power $P_{leakage}$. P_{sw} is often the most significant source, therefore efforts to reduce the power consumed in FIR system realizations, focus on reducing P_{sw} . Since multiplications represent the most complex task in FIR computations, a lot of research has been carried out on reducing the complexity of or totally eliminating multiplications in computing the product terms in Eqn. (1) [10].

FIR filters have a large number of multiplications involved in the filter algorithm, which are usually implemented in floating point arithmetic (IEEE 754 double-precision binary floating-point format: binary32).

The floating point number system can accommodate a large range of numbers and so in floating point arithmetic higher accuracy in processing can be achieved. But the hardware implementation for floating point arithmetic is costlier and the speed of processing is low due to double calculations i.e., separate calculation for mantissa and exponent. In this arithmetic, the truncation and rounding errors occur both for multiplication and addition, whereas in fixed point arithmetic such errors occur only for multiplication. The addition in fixed point arithmetic leads to overflow, but the overflow is rare phenomena in floating point arithmetic due to larger dynamic range. Therefore, the floating point arithmetic is preferred for non-real time applications on general purpose systems (computers) in which the cost and speed are not significant and fixed point arithmetic is preferred due to the reduced cost of the hardware and high speed processing [12].

V. MULTI-OBJECTIVE PROBLEM FORMULATION

Multi-objective optimization involves minimizing or maximizing multiple objective functions subject to a set of constraints. Example problems include analyzing design tradeoffs, selecting optimal product or process designs, or any other application where you need an optimal solution with tradeoffs between two or more conflicting objectives.

In VLSI implementation of digital FIR filters, design constraints which influence the performance of FIR filters are area, power and delay. But main hurdle in VLSI implementation of digital FIR system is that either design can be area efficient or power efficient or speed efficient; but not all area-time-speed efficient simultaneously. Optimizing one parameter affects the others.

So, the objective of this research work is to come up with step by step an optimal multi-objective approach for VLSI implementation of digital filters wherein all constraints viz. area, power and time are optimized simultaneously.

VI. DESIGN APPROACH

A. Obtain Transposed Form of FIR Filter of figure 1.1.

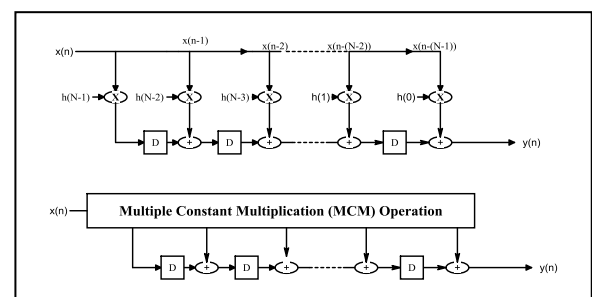


Fig. 5.1.: FIR Filter Transposed Form

Advantages of transposed form are:-

- Computationally equivalent to direct form.
- Can be obtained by reversing order of final addition followed by retiming. Now, all multiplications share one input.
- The direct-form structure has the disadvantage that each adder has to wait for the previous adder to finish

before it can compute its result. For high speed hardware such as FPGAs/ASICs, this introduces latency which limits how fast the filter can be clocked. A solution to this is to use the transposed direct-form structure instead. With this structure, the delays between the adders can be used for pipelining purposes and therefore all additions/multiplications can be performed in fully parallel fashion. This allows real-time handling of data with very high sampling frequencies and also provides a solution to optimize the speed of the system.

VII. MULTIPLICATION

Multiplication in digital FIR designs often involves the multiplication by constant coefficients as shown in figure 1.1. The shift and add loop of traditional multipliers can be replaced with a set of high speed wire-shifts and then added in one quick step while still fulfilling the same binary multiplication shown in Equation 2.1.

$$k = \sum_{i=0}^n 2^i k_i; \quad (2.1)$$

This optimization is sometimes referred to as “multiplierless” design, although the shift and add structure created does still implement a multiplier. Single constant multiplication (SCM) is also a term that is used to describe the optimized constant multipliers. In hardware, the multiplication operation is considered to be expensive, as it occupies significant area. Hence, constant multiplications are generally realized using only addition, subtraction, and shift operations [5].

The logic for obtaining the shift and add structure of an SCM (figure 5.2) is to first convert the constant multiplicand into its binary form. For example, the constant $(43)_{10}$ is converted to $(101011)_2$. Then to multiply x by $(43)_{10}$, shift x by a set amount for each 1 digit in the binary encoding. The amount of the shift is determined by the order of magnitude of that particular bit position. For $(43)_{10}$, the MSB of the binary encoding is a 1, so x needs to be shifted left by five because the MSB has the magnitude of 32. The final step is to then add all of shifted values to compute the product.

The number of 2-input additions necessary to perform the constant multiplication is the number of nonzero digits of the binary representation minus one. The example coefficient $(43)_{10}$, $(101011)_2$, requires three adders to form a product because there are four nonzero digits. While this optimization for constant multiplications is useful, it is not optimal.

The multiplier block of the digital FIR filter in its transposed form [Fig. 5.1], where the multiplication of filter coefficients with the filter input is realized, has significant impact on the complexity and performance of the design because a large number of constant multiplications are required. This is generally known as the multiple constant multiplications (MCM) operation. The goal is the minimization

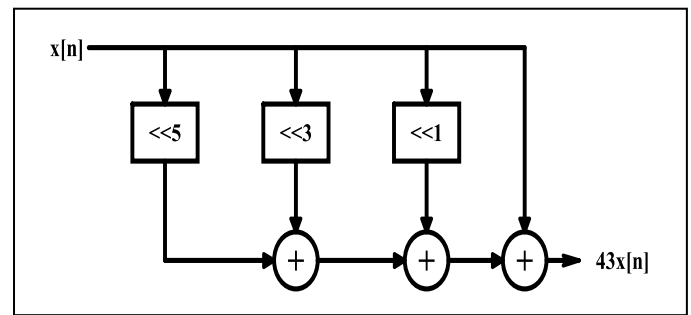


Figure 5.2: Example of SCM approach based multiplier design

of nonzero terms within the discrete coefficients as each nonzero term corresponds to an additional adder in the hardware implementation. Depending on the target hardware, it may be possible to implement a linear-phase FIR filter using less multipliers than the minimum-phase filter by taking advantage of the symmetry even if the filter length of the linear-phase is larger [3,4].

For the bit-parallel design of the MCM operation, the MCM problem is defined as finding the fewest number of addition and subtraction operations that realize the MCM, since shifts can be implemented using only wires in hardware.

Many efficient algorithms [6, 7] have been introduced for the MCM problem. In spite of various methods they use and different search space they explore, the main idea has always been the maximization of the sharing of common partial products among the constant multiplications. As an example, consider the constant multiplications $29x$ and $43x$. Observe from Figure 5(a)-(b) that the sharing of partial products $3x$ and $5x$ reduces the number of operations from 6 to 4. The same sharing of partial products approach has been used in our transposed form structure [11]. Thus, when using MCM instead of SCM, an added savings can be accomplished by reusing fundamentals between the constants.

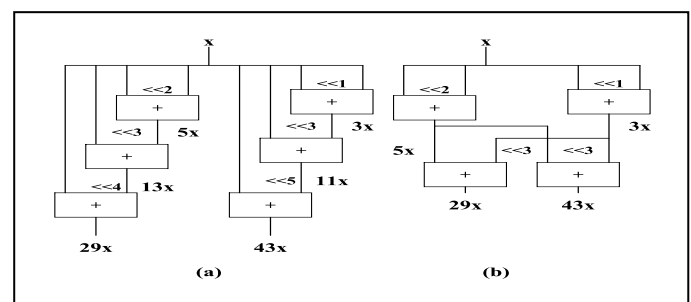


Fig. 5.3: Shift-adds implementations of $29x$ and $43x$ (a) without partial product sharing; (b) with partial product sharing.

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VIII. COMPARISON AND SIMULATION RESULT

In our work, we had designed three LTI filters viz. filter 1 (direct form), filter 2 (transposed form) and filter 3 (Optimized direct form) and then their performance was compared with respect to area, dynamic power dissipation and propagation delay.

Firstly, simple direct form FIR filter structure was implemented in MATLAB using FDA (Filter Design & Analysis) tool of MATLAB with following specifications:-

- Design Method :- FIR equiripple
- Response type :- Low pass
- Filter order:- 17

Elaborated settings done in FDA tool are shown in figure below:-

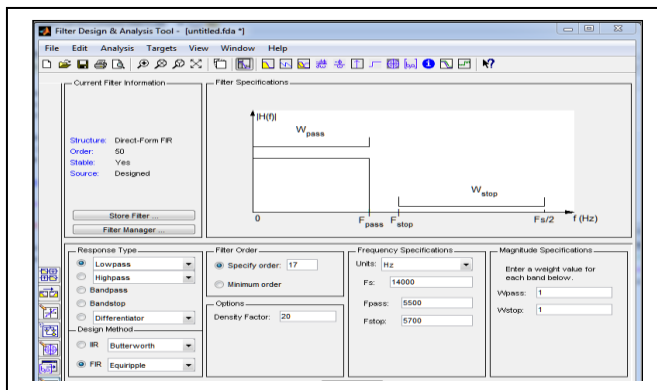


Fig.6.1: FIR Equiripple filter specification on FDA tool

From FDA tool, the filter co-efficients for direct form FIR filter structure were obtained. But these co-efficients were negative and in floating point format. In order to optimize the resources used (i.e. gates and hence area) at RTL (Register Transfer level) , processing performance, system cost and ease of use; and since dynamic range of output is known, the floating point coefficients were converted into the fixed point coefficient by multiplying them with 1000 and taking the round off value of it. After that negative coefficients were converted into the positive coefficients by taking the absolute value of previous value.

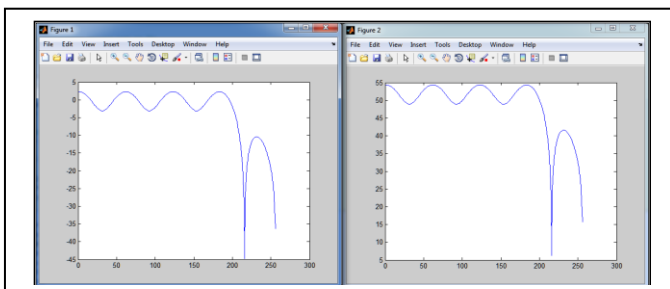


Fig.6.2 Magnitude Response of FIR Filter (a) Using floating point arithmetic (b) Using fixed point arithmetic

The magnitude response of direct form FIR filter using floating point arithmetic and fixed point arithmetic were found to be same (figure 6.2).

Next, using these filter co-efficients direct form, transposed form and optimized direct form FIR filter structure were implemented using Active HDL and their performance with respect to area, timing and dynamic power consumption were analysed using Xilinx tool at RTL level and Cadence SOC encounter tool at Layout level.

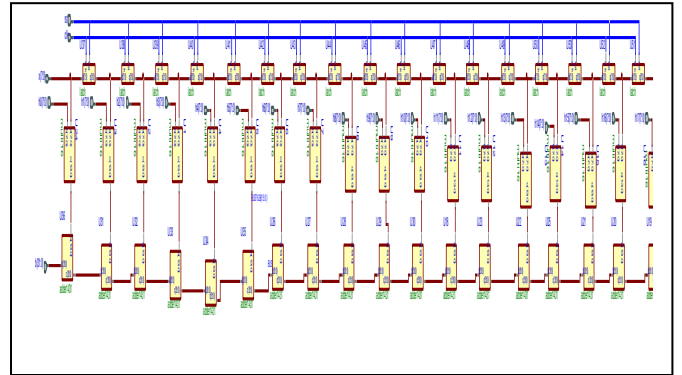


Fig.6.3: Direct Form FIR filter using Active HDL

The multiplier unit of MAC in direct form FIR filter is implemented using generic multiplier. The MCM block of MAC unit in transposed form is implemented using structure shown in figure 5.3(b). As per the concept of MCM approach derived from SCM approach, number of adders required to implement MAC unit of transposed structure will be high and thereby affecting area and power consumption of transposed form FIR filter structure.

To optimize the problems faced in implementing the MAC unit of transposed form digital filter structure, we had suggested a slight modification in transposed form FIR filter structure and with same approach DF FIR was also redesigned. In optimized transposed form structure concept of co-efficient reuse is used to optimize further the resources used in comparison to structure shown in figure 5.3(b) and direct form structure FIR filter structure. In other words, it was observed that from 18 filter co-efficients obtained from FDA tool of MATLAB, five co-efficients were repeated two or three times. So we had designed only five multiplier unit based on structure shown in figure 5.3(b). The final direct form structure of FIR filter and transposed of direct form structure of FIR filter are shown in figure 6.3 and figure 6.4 respectively. The optimized DF form structure is shown in figure 6.5.

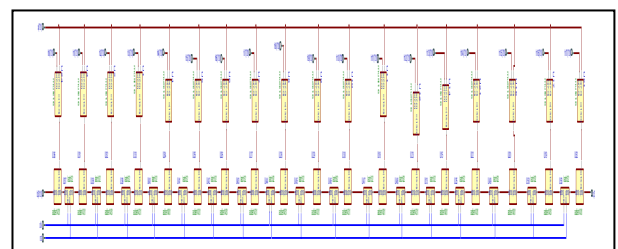


Fig.6.4: Transposed of Direct Form FIR filter using Active HDL

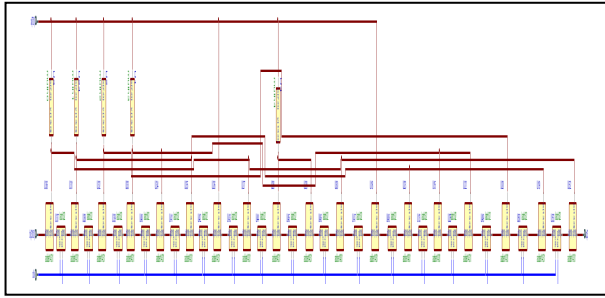


Fig.6.5: Optimized Direct Form FIR filter using Active HDL

Now, on comparing direct form FIR structure and transposed form FIR structure, it was observed that latch size in transposed form at each stage is increasing and hence the adder size is also increasing. So, this may increase area overhead and hence power consumption of design and also may add to the latency of the circuit. The same was observed after the implementation of transposed structure.

Compilation Report of direct form (filter 1), transposed form (filter 2) and optimized direct form (MCM with partial product sharing) (filter 3) of FIR filter are given below:-

Table 7.3.1 Compilation Summary of Filter 1, Filter 2 & Filter 3

	Filter 0	Filter 1	Filter 2
Technology	gsc145nm	gsc145nm	gsc145nm
Global Operating Voltage	1.1 V	1.1 V	1.1 V
Combinational area	27431.992899 sq.nm	28421.277155 sq.nm	14868.831567 sq.nm
Noncombinational area	1404.145630 sq.nm	5616.582520 sq.nm	5620.336920 sq.nm
Total cell area	28836.138528 sq.nm	34037.859675 sq.nm	20489.168487 sq.nm
Cell Internal Power	3.5082 mW	7.7709 mW	1.4805 mW
Net Switching Power	2.3820 mW	5.0423 mW	864.7971 uW
Total Dynamic Power	5.8902 mW	12.8132 mW	2.3453 mW
Cell Leakage Power	172.0155 uW	225.0586 uW	152.1185 uW
Worst Case Propagation delay (RTL Xilinx report)	21.101 nsec	11.129 nsec	6.619 nsec

IX. CONCLUSION

In this paper, implementation of low power, speed efficient and area efficient FIR filters using filter co-efficient reuse concept and MCM technique based on partial product sharing has been considered wherein multiplication operations are replaced by shift-and-add operation. The experimental results showed that area has been reduced by 28.946 %, dynamic power consumption reduced by 60.183% and worst propagation delay by 68.232 % in optimized direct form FIR filter structure in comparison to direct form and transposed form digital FIR filter structure. This indicated that our proposed modification in direct form FIR filter structure and use of MCM technique based on partial product sharing and

use of concept co-efficient sharing leads to an area efficient, low power and high speed digital FIR Filter structure for DSP systems.

Future research includes improving the performance of the FIR system by implementing if possible adder unit using fast adders and a full characterization of each design option at layout level.

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