# High Speed Area Efficient Radix-2 Fast Fourier transforms using Signed and Complex Number

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**Abstract:** with the advent of new technology in the fields of VLSI and communication, there is also an ever growing demand for high speed processing and low area design. It is also a well-known fact that the chip area and maximum combinational path delay (MCPD) unit forms an integral part of processor design. Due to this regard, high speed and low area architectures become the need of the day. A fast Fourier transform (FFT) is any fast algorithm for computing the DFT. The development of FFT algorithms had a tremendous impact on computational aspects of signal processing and applied science. The decimation-in-time (DIT) fast Fourier transform (FFT) very often has advantage over the decimation-in-frequency (DIF) FFT for most real-valued applications, like speech/image/video processing, biomedical signal processing, and time-series analysis, etc., since it does not require any output reordering.

Index Terms: FFT, Decimation in Time, Decimation in Frequency, real Value data

## I. INTRODUCTION

The discrete Fourier transform (DFT) is an important tool in many branches of science and engineering [1] and has been studied extensively [2]. For many practical applications, it is important to have an implementation of the DFT that is as fast as possible. In the past, speed was the direct consequence of clever algorithms [2] that minimized the number of arithmetic operations. On present day general-purpose microprocessors, however, the performance of a program is mostly determined by complicated interactions of the code with the processor pipeline, and by the structure of the memory. Designing for performance under these conditions requires an intimate knowledge of the computer architecture. In this paper, we address this problem by means of a novel adaptive approach, where the program itself adapts the computation to the details of the hardware. We developed FFTW, an adaptive, high performance implementation of the Cooley-Tukey fast Fourier transform (FFT) algorithm [3], written in C. We have compared many C and FORTRAN implementations of the DFT on several machines, and our experiments show that FFTW typically yields significantly better performance than all other publicly available DFT software.

The FFT (Fast Fourier Transform) and its inverse (IFFT) are the key components of OFDM (Orthogonal Frequency Division Multiplexing) systems. Recently, the demand for long length, high-speed and low-power FFT has increased in the OFDM applications. There are three kinds of main design architectures for implementing a FFT processor. One is the single-memory architecture. It has one processing element and one main memory. Hence, it occupies a small area. The second is the dual memory architecture, which has two memories. This architecture has a higher throughput than the single-memory architecture because it can store butterfly outputs and read butterfly inputs at the same time. The fast Fourier transform plays an important role in many digital signal processing (DSP) systems. Recent advances in semiconductor processing technology have enabled the deployment of dedicated FFT processors in applications such as telecommunications, speech and image processing. Specifically, in the OFDM communication systems, FFT and inverse FFT (IFFT) play a very important role. The OFDM technique, due to its effectiveness in overcoming adverse channel effects [1, 2] as well as spectrum utilization, has become widely adopted in wire line and wireless communication standards. The OFDM technique has been adopted in several standards like digital audio broadcasting (DAB) [3], digital video broadcasting terrestrial (DVB-T) [4], asymmetrical digital subscriber line (ADSL) [5] and very-highspeed digital subscriber line (VDSL) [6]. Therefore, efficient and low-power VLSI implementation of FFT processors is essential for successful deployment of these OFDM-based systems. According to the standards of DAB, DVB-T, ADSL and VDSL, various FFT sizes are required, as shown in Table 1. From this Table, it is clear that variable-length FFT hardware is a crucial module in the low-cost solution of the above communication systems. The Cooley – Tukey N-point FFT algorithm requires O (Nlog N) computations, which is a huge saving over direct computation of the discrete Fourier transform (DFT). However, hardware implementation of the algorithm is both computational intensive, in terms of arithmetic operations, and communication intensive, in terms of data swapping. For real-time processing of FFT, O(log N) arithmetic operations are required per sample cycle. High speed real-time processing can be accomplished in two different ways. In the conventional general-purpose digital signal processor (DSP) approach, the computation is carried out by a single processor

driven to a high clock frequency, which is O(log N) times the data sample frequency. In the application specific parallel or pipelined processor approach, the required operations are performed at the clock frequency equivalent to the sample frequency, and this approach usually consumes less power.

#### II. FAST FOURIER TRANSFORM

Before going further to discuss on the FFT and IFFT design, it is good to explain a bit on the fast Fourier transform and inverse fast Fourier transform operation. The fast Fourier transform (FFT) and inverse fast Fourier transform (IFFT) are derived from the main function which is called Discrete Fourier Transform (DFT). The idea of using FFT/IFFT instead of DFT is that the computation of the function can be made faster where this is the main criteria for implementation in the digital signal processing. In DFT the computation for N-point of the DFT will calculate one by one for each point. While for FFT/IFFT, the computation is done simultaneously and this method saves quite a lot of time. Below is the equation (2.2) showing the DFT and from here the equation is derived to get FFT/IFFT function.

$$X(k) = \sum_{n=0}^{N-1} x(n) e^{-j2\pi k / N}$$
(1)

X (k) represent the DFT frequency output at the *k*-the spectral point where *k* ranges from 0 to N-1. The quantity N represents the number of sample points in the DFT data frame.

The quantity x (n) represents the *n*th time sample, where *n* also ranges from 0 to *N*-1. In general equation, x (n) can be real or complex.

The DFT equation can be re-written equation (2) into:

$$X(k) = \sum_{n=0}^{N-1} x(n) W_N^{nk}$$
(2)

The quantity  $W_N^{nk}$  is defined as in equation (3)

$$W_N^{nk} = e^{-j2\pi k/N} \tag{3}$$

Here is where the secret lies between DFT and FFT/IFFT where the equation (2.4) function above is called Twiddle Factor. This factor is calculated and put in a table in order to make the computation easier and can run simultaneously. The Twiddle Factor table is depending on the number of point use. During the computation of IFFT, the factor does not to recalculate since it can refer to the Twiddle factor table thus it save time since calculation is done concurrently.

#### **Inverse Fast Fourier Transform**

Inverse fast Fourier transform (IFFT) is used to generate OFDM symbols. The data bits is represent as the frequency domain and since IFFT convert signal from frequency domain to time domain, it is used in transmitter to handle the process.

**Table 1:** Twiddle factor for 8 point inverse fast Fourier transformIFFT (N=8)

No.	W	Value					
1	$W_8^{-0}$	1					
2	$W_8^{-1}$	0.7071+j0.7071					
3	$W_{8}^{-2}$	j					
4	$W_{8}^{-3}$	-0.7071+j0.7071					
5	$W_8^{-4}$	-1					
6	$W_{8}^{5}$	-0.7071-j0.7071					
7	$W_{8}^{-6}$	-j1					
8	$W_{8}^{-7}$	0.7071-j0.7071					

IFFT is defined as the equation (4) below:

$$x(n) = \frac{1}{N} \sum_{n=0}^{N-1} X(k) W_N^{-nk}$$
(4)

Same FFT algorithm can be used to find IFFT function with the changes in certain properties. The changes that implement is by adding a scaling factor of 1/N and replacing twiddle factor value () with the complex conjugate () to the above equation. With these changes, the same FFT flow graph also can be used for the inverse fast Fourier transform. Below is the table 2.2 show the values of twiddle factor for IFFT.



Figure 1: Radix-2 Decimation in Time Domain FFT Algorithm

#### III. PROPOSED METHOD

The flow chart of the proposed methodology is shown in figure 3. In this paper we are used three techniques i.e. unsigned multiplier, signed multiplier and complex multiplier. In this figure the two signed and unsigned bit multiplier (i.e. multiplier n-bit, multiplicand m-bit) and final output of the multiplier is n+m bits. All the multiplier is design into two parts i.e. partial product generator and multi operand addition.



Figure 3: Flow Chart of the proposed Methodology

Array Multiplier:-

Signed Multiplier:-



**Complex Multiplier:-**



The FFT algorithms are classified into two broad categories, namely, the decimation-in-time (DIT) and the decimation-infrequency (DIF) algorithms. The proposed radix-2 DIT algorithm is shown in figure 3.



DIT butterfly involves a multiplication followed by additions. As shown in Table I the computation time of fixed-point multiplication followed by an addition is less than that of addition followed by a multiplication. The DIT-based FFT butterfly thus involves less propagation delay than that of DIF-based RFFT butterfly although both these butterflies involve the same number of multipliers and adders. Therefore, the choice of DIT algorithm to derive FFT structure has an advantage over DIF algorithm. In this paper, we present an efficient architecture for the DIT radix-2 RFFT algorithm. Derivation of proposed RFFT structure using register-based storage.



Figure 6: Proposed 16-point Radix-2 DIT Algorithm

This algorithm decomposes a sequence of DFT into four small DFTs of 1/4 lengths in a recursive manner and their outputs are employed to manipulate several other outputs by which the cost of computation will be reduced. The input data is disintegrated into four small sequences of x (4n + i) where n = 0, 1... N/4-1 and i = 0, 1, 2, 3.

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Figure 7: Proposed 32-point Radix-2 DIT Algorithm

This algorithm decomposes a sequence of DFT into four small DFTs of 1/3 lengths in a recursive manner and their outputs are employed to manipulate several other outputs by which the cost of computation will be reduced. The input data is disintegrated into four small sequences of x (3n + i) where n = 0, 1... N/3-1 and i = 0, 1, 2.



Figure 9: Proposed 9-point Radix-3 DIT Algorithm

# IV. DELAY AND AREA METHODOLOGY

The delay and area evaluation methodology considers all gates to be made up of AND, OR, and Inverter (AOI), each having delay equal to 1 unit and area equal to 1 unit. We then add up the number of gates in the longest path of a logic block that contributes to the maximum delay. The area evaluation is done by counting the total number of AOI gates required for each logic block.

Adder Block	Delay	Area
XOR	3	5
2:1 MUX	3	4
Half Adder	3	6
Full Adder	6	13

Table 1: Delay and Area calculate in basic block of FFT
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Table 2: Computational Delay of Mult-add And Add-mult Operations

Architecture	T <sub>MA</sub>	T <sub>AM</sub>	$T_{AM}$ - $T_{MA}$	% Difference
8-bit				
Pramod Kumar Meher et al. [1]	8.345	8.967	0.622	6.9%
Proposed Design	8.048	8.343	0.295	3.5%
16-bit				
Pramod Kumar Meher et al. [1]	9.453	10.321	0.868	8.4%
Proposed Design	8.653	9.240	0.587	6.3%
32-bit				
Pramod Kumar Meher et al. [1]	14.532	14.982	0.45	3.0%
Proposed Design	13.377	14.010	0.633	4.5%

# V. SIMULATION RESULT

All the designing and experiment regarding algorithm that we have mentioned in this paper is being developed on Xilinx 6.2i updated version. Xilinx 6.2i has couple of the striking features such as low memory requirement, fast debugging, and low cost. The latest release of  $ISE^{TM}$  (Integrated Software Environment) design

tool provides the low memory requirement approximate 27 percentage low. ISE 6.2i that provides advanced tools like smart compile technology with better usage of their computing hardware provides faster timing closure and higher quality of results for a better time to designing solution. ISE 6.2i Xilinx tools permits greater flexibility for designs which leverage embedded processors. Also included is the newest release of the chip scope Pro Serial IO Tool kit, providing simplified debugging of high-speed serial IO designs for Virtex-2p FX and Virtex-2p LXT and SXT FPGAs. With the help of this tool we can develop in the area of communication as well as in the area of signal processing and VLSI low power designing.

Radix-2 DIT Algorithm for N=8						
Parameter	Number of Slice	Number of LUTs	MCPD (nsec)			
Proposed Design (unsigned)	96	192	15.548			
Proposed Design (signed)	221	387	15.137			
Radix-2 DIT Algorithm for N=1	6	·				
Parameter	Number of Slice	Number of LUTs	MCPD (nsec)			
P. K. Meher et al. [1]	-	-	21.054			
Proposed Design (unsigned)	256	512	19.260			
Proposed Design (signed)	573	1000	18.927			
Radix-2 DIT Algorithm for N=3	52					
Parameter	Number of Slice	Number of LUTs	MCPD (nsec)			
P. K. Meher et al. [1]	-	-	24.421			
Proposed Design (unsigned)	640	1280	22.147			
Proposed Design (signed)	1410	2462	22.806			
Radix-2 DIT Algorithm for N=6	54					
Parameter	Number of Slice	Number of LUTs	MCPD (nsec)			
P. K. Meher et al. [1]	-	-	29.421			
Proposed Design (unsigned)	1536	3072	26.319			
Proposed Design (signed)	2431	4063	22.785			
Radix-2 DIT Algorithm for N=1	28	·				
Parameter	Number of Slice	Number of LUTs	MCPD (nsec)			
P. K. Meher et al. [1]	-	-	36.421			
Proposed Design (unsigned)	2739	5042	32.895			
Proposed Design (signed)	3390	6042	33.806			

Table III: Comparison result for existing algorithm and proposed algorithm

#### Table IV: Result for complex number

Radix-2 DIT Algorithm for N=8 and N=16					
Parameter Number of Slice Number of LUTs MCPD (nsec)					
N=8	438	864	24.948		
N=16	1168	2304	30.687		



Slice LUTs

**Figure 9:** Bar graph of the complex number for N=8 and N=16

Table	<b>X7T</b> .	D	. <b>f</b>	DIT	and DI	$\mathbf{T} = 1$		£		al a a
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Parameter	Number of Slice	Number of LUTs	MCPD (nsec)
DIT algorithm for N=9	144	288	15.109
DIF algorithm for N=9	120	240	15.640



Figure 10: Bar graph of the radix-3 algorithm for N=9	
<b>Table V:</b> Result for DIT and DIF algorithm for radix-4 algorithm	n

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Parameter	Number of Slice	Number of LUTs	MCPD (nsec)			
DIT algorithm for N=16	320	640	18.526			
DIF algorithm for N=16	289	568	19.095			



**Figure 11:** Bar graph of the radix-4 algorithm for N=16

# VI. CONCLUSION

The prime objective is to construct a FFT in order to have low power consumption and lesser area. The parameters (i) power consumption (ii) Area occupancy were given due consideration for comparing the proposed circuit with other FFTs. The circuits were simulated using Model-Sim 6.3c and synthesized with Xilinx ISE 14.1.The performance of various 64 point FFT such as Radix-2, Radix-4, split Radix, mixed -radix 4-2, R2MDC and the proposed modified R2MDC were carried out and their performance were analyzed with respect to the number of CLB slices, utilization factor and Power consumption.

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