

FPGA Implementation of the Fast Fourier Transform

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Introduction

In this project we implemented a fast Fourier transform algorithm on a Xilinx FPGA using VHDL. The primary motivation of our project was to digitally analyze musical signals, but the use cases for an FFT chip far surpass that specific use case.

As a very general overview, our completed device functions by sampling an analog signal for a set amount of time and storing the data into an input buffer in BRAM. Next, the signal is processed using an FT controller and a complex ALU and stored in an output buffer. Finally, a microcontroller can then read the output buffer over a standard SPI protocol. From there the processed frequency domain data can be sent to a computer for further processing or any other device.

Theory

To understand the discrete Fourier transform, one must first analyze its analogous continuous time form. This is written most simply as

$$X(\omega) = \int_{-\infty}^{\infty} x(t)e^{-j\omega t} dt$$

which defines the transform of a continuous time signal $f(t)$. Without getting into too many details, it defines the signal as a combination of sinusoids, making it a very useful tool for real world applications.

The discrete Fourier transform (DFT) is simply a reduction of the continuous Fourier transform into a discrete sample space. In other words, if we let x_n represent a sampled version of the continuous time function $x(t)$ with a total of N samples, we can replace the integral with a summation over the series, as shown below.

$$X_k = \sum_{n=0}^{N-1} x_n e^{\frac{-j2\pi kn}{N}}$$

As with the continuous time case, this series gives us a glimpse into the component elements of our original signal. However, it is rather costly to

compute, requiring a time complexity of $O(N^2)$. For each value X_k , a series of values from $n = 0$ to $N - 1$ must be generated and summed, using up valuable computer resources. Thus, calculating the DFT in this way is very inefficient, so we instead turned to the fast Fourier transform.

In order to perform this operation more quickly, we utilized the Cooley-Tukey FFT algorithm. However, the one requirement is that our input is strictly a power of 2. That is because the algorithm works by recursively finding the FFT of smaller and smaller sample sizes of x_n , which arises from the fact that the transform itself is periodic. The transform function can be broken into the sum of its even and odd components,

$$X_k = \sum_{m=0}^{N/2-1} x_{2m} e^{-\frac{2\pi i}{N}(2m)k} + \sum_{m=0}^{N/2-1} x_{2m+1} e^{-\frac{2\pi i}{N}(2m+1)k}$$

This equation can further be simplified by making the substitutions

$$E_k = \sum_{m=0}^{N/2-1} x_{2m} e^{-\frac{2\pi i}{N}(2m)k} \quad O_k = \sum_{m=0}^{N/2-1} x_{2m+1} e^{-\frac{2\pi i}{N/2}mk}$$

giving us

$$X_k = E_k + e^{-\frac{2\pi i}{N}k} O_k$$

The expression $e^{-\frac{2\pi i}{N}k}$ is commonly called the *twiddle factor*. Furthermore, because of the periodicity of the transform, we can calculate respective even and odd components simultaneously

$$E_k = E_{k+\frac{N}{2}} \quad O_k = O_{k+\frac{N}{2}}$$

Stuff That's Already Been Done (change section name plz)

Motivation

System Overview

A critical task in building our FFT device was converting an arbitrary waveform into the SPI signal input that the FPGA expected. This requires the

use of a fast analog to digital converter. The Xilinx Spartan 3E FPGA does not come with an onboard ADC. Because of this, a large percentage of the work was actually electronics work done outside of the FPGA:

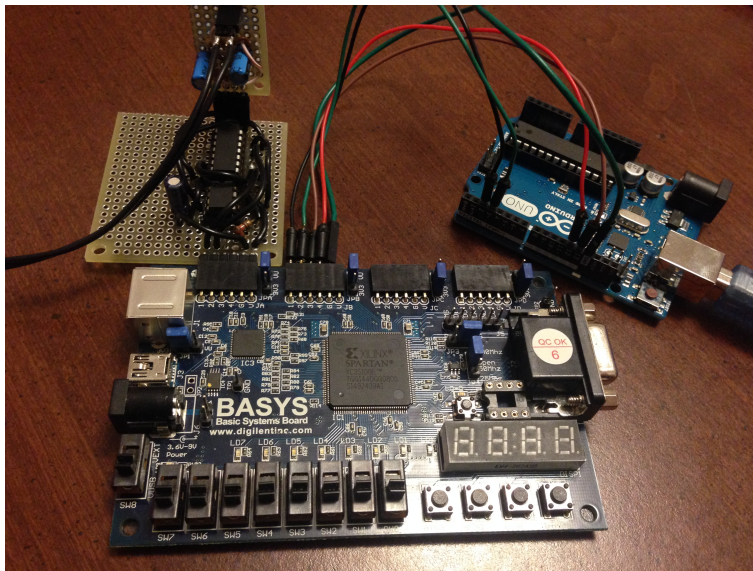


Figure 1: Full system including ADC PMOD and audio signal module.

These external components condition the signal and then send the digital signal into the FPGA at the right voltage level. In addition, the FPGA's output buffer must be read out by an Arduino microcontroller and sent to a computer with the right software to parse and display the output. We call these external components macro-components and we call the FPGA internal blocks micro-components.

Macro-Component Descriptions

The required components external to the FPGA are as follows:

Audio signal module

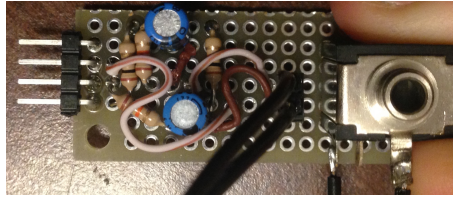


Figure 2: Audio signal circuit board with audio connector

The audio signal module tailors the input audio signal for conversion with the ADC. Traditional audio signals are centered around 0 V and are in the negative voltage range half the time. This module biases the signal by 2.5 V, allowing for a 5 V variation peak to peak in the audio signal without any clipping. This module also provides filtering capacitors that remove most of the supply rail noise that may be present.

ADC PMOD

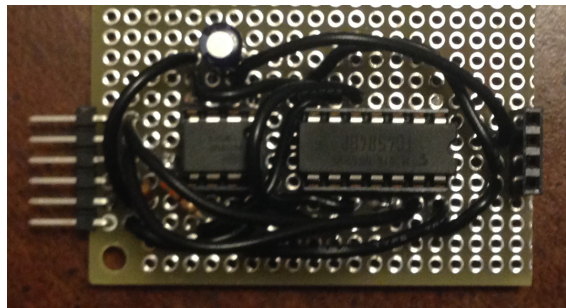


Figure 3: ADC PMOD circuit board

The ADC PMOD accepts any input analog input and converts it to a SPI bus that the FPGA can read. The analog signal is fed into the onboard ADS7818 chip. Unfortunately, this chip operates at 5 V but the FPGA operates at 3.3 V. Because of this, there is a voltage divider to drop the voltage down to the FPGA. For FPGA output to be read by the ADS7818, we added a Schmitt inverter IC wired as a Schmitt trigger to act as a level converter.

Arduino

The Arduino functions as a means of getting the processed information out of the FPGA's BRAM. The Arduino turned out to be the main bottleneck in terms of framerate to the computer's display. We had to push the SPI and serial write functions outside of recommended bounds to get a fluid framerate on the computer display. The full software is available in the `arduino/osc.binary_spi` directory of the source code repository.

Computer software

The computer software for this project is divided into two components. The first component is the Python module that facilitates the reading and decoding of the data from the serial port. It will read out 512 complex numbers and, optionally, compute the magnitude of each of those numbers. The second component is the actual display. It will use the Python module to read the device and it uses Tkinter to display the plot on the screen.

Micro-Component Descriptions

The final block diagram of our device is as follows:

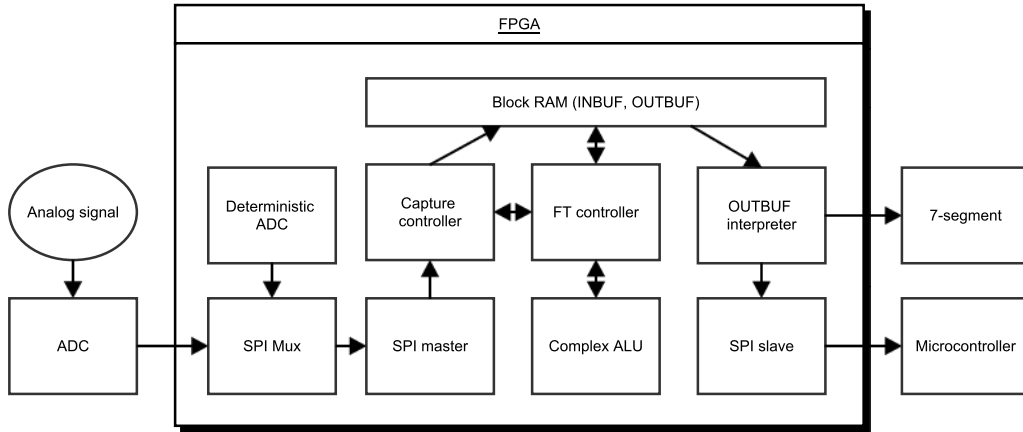


Figure 4: Block diagram of the FFT chip

All of the blocks inside the FPGA device are referred to as micro-components.

TODO: DESCRIPTION OF EACH COMPONENT HERE

Steps Taken

Simulation Results

Results

Roadblocks

Future Improvements

Conclusion

Appendix

References

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