

## Report for Lab #4: Using FFT for Frequency Analysis

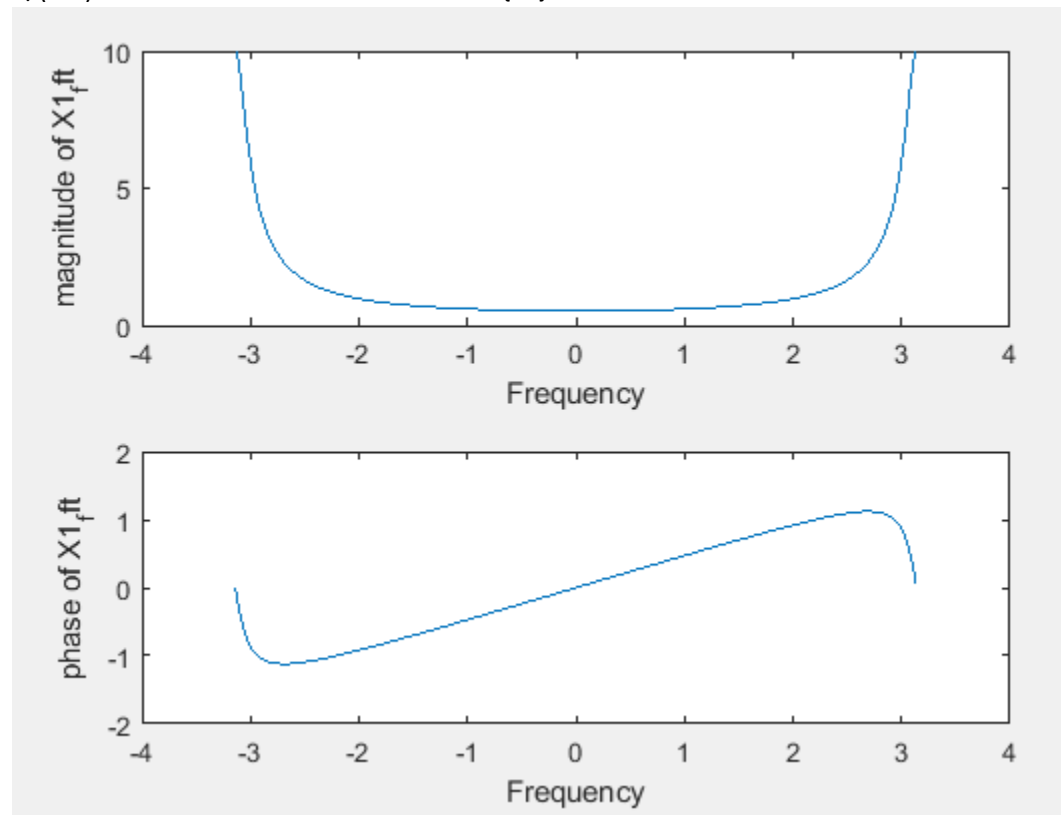
Report by:  
Luke Jiang 1560831

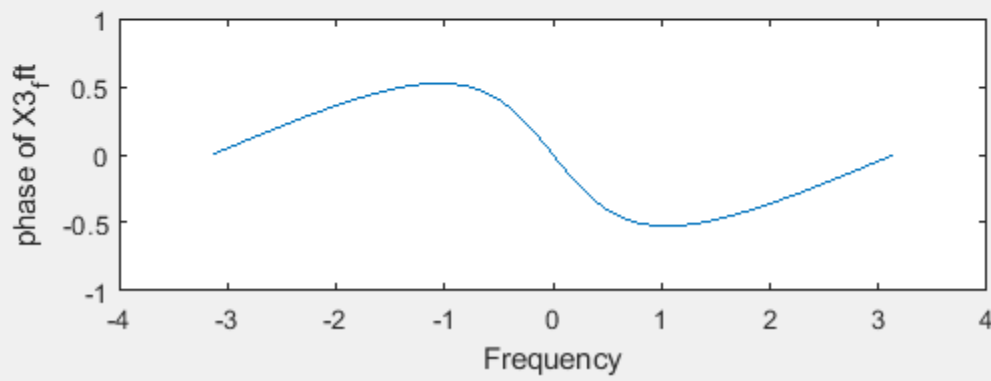
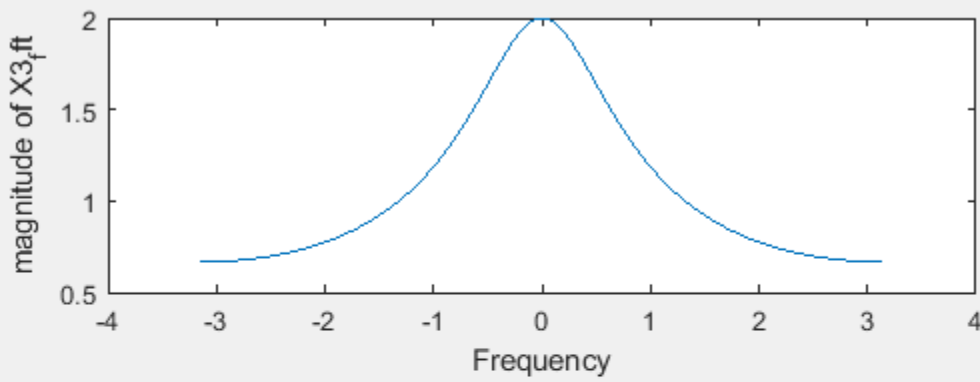
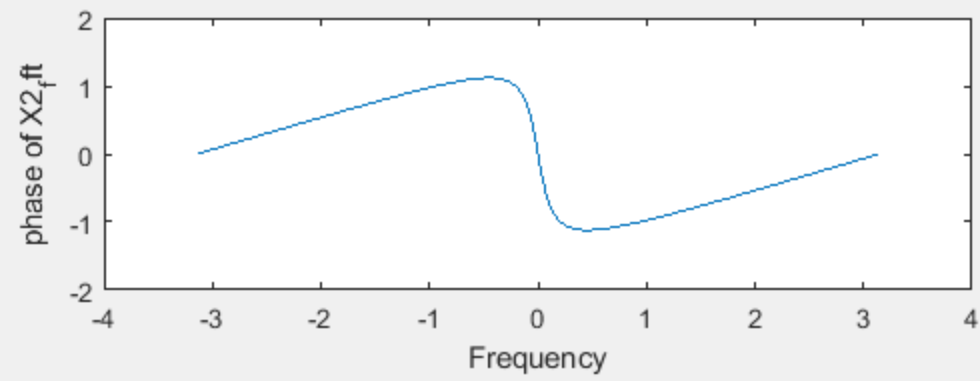
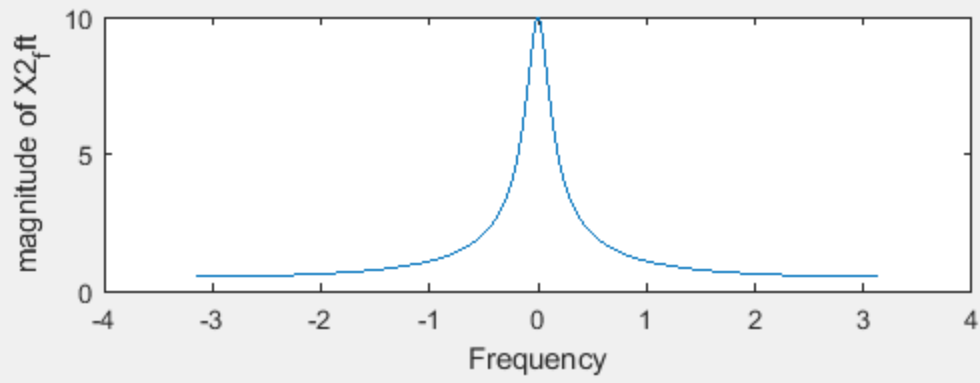
### 1 ASSIGNMENT I

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Q1: Turn in the plots for the different signals. Specify the window length  $N$  and explain why the DFTs look different in terms of Fourier Transform properties.

A1: Since  $x_1[n] = (-1)^n * u[n] = e^{j\pi n} * u[n]$ ,  $\text{FT}\{x_1[n]\}$  is a  $2\pi$ -shifted version of  $\text{FT}\{x_2[n]\}$  by time-shifting property. Also, since  $x_3[n]$  is narrower than  $x_2[n]$  in time domain,  $\text{FT}\{x_3[n]\}$  is wider than  $\text{FT}\{x_2[n]\}$  in frequency domain by inverse time-frequency relationship. The magnitude of  $\text{FT}\{x_3\}$  is also smaller by equation  $X(e^{j\omega}) = 1/(1 - a * e^{j\omega})$ . At  $\omega=0$ ,  $X(1) = 1/(1-a)$ . if  $a$  decreases,  $1-a$  increases and  $1/(1-a)$  decreases. So the max value of  $\text{FT}\{x_3\}$  decreases.





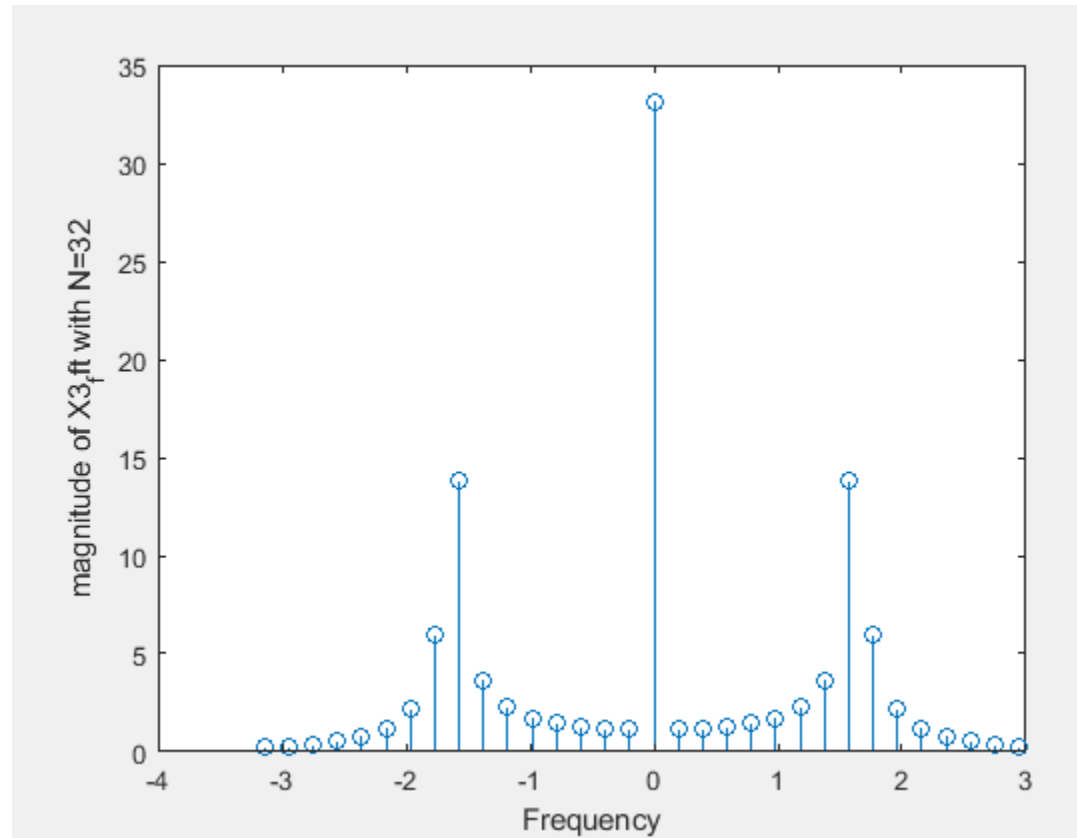
## 2 ASSIGNMENT II

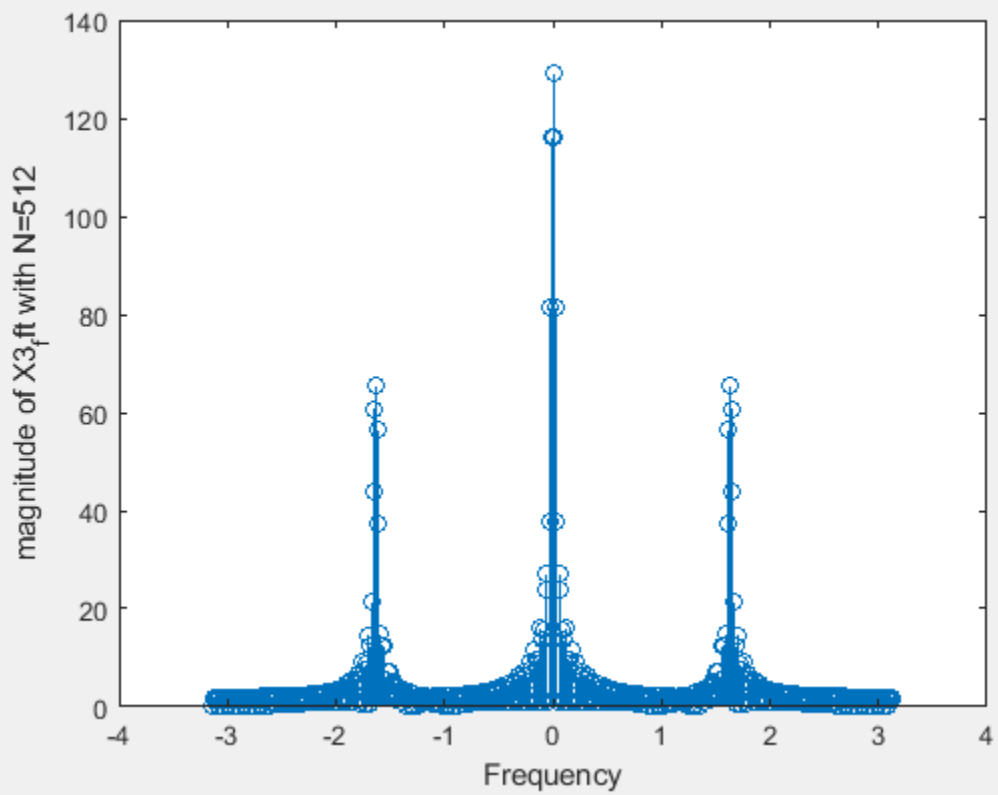
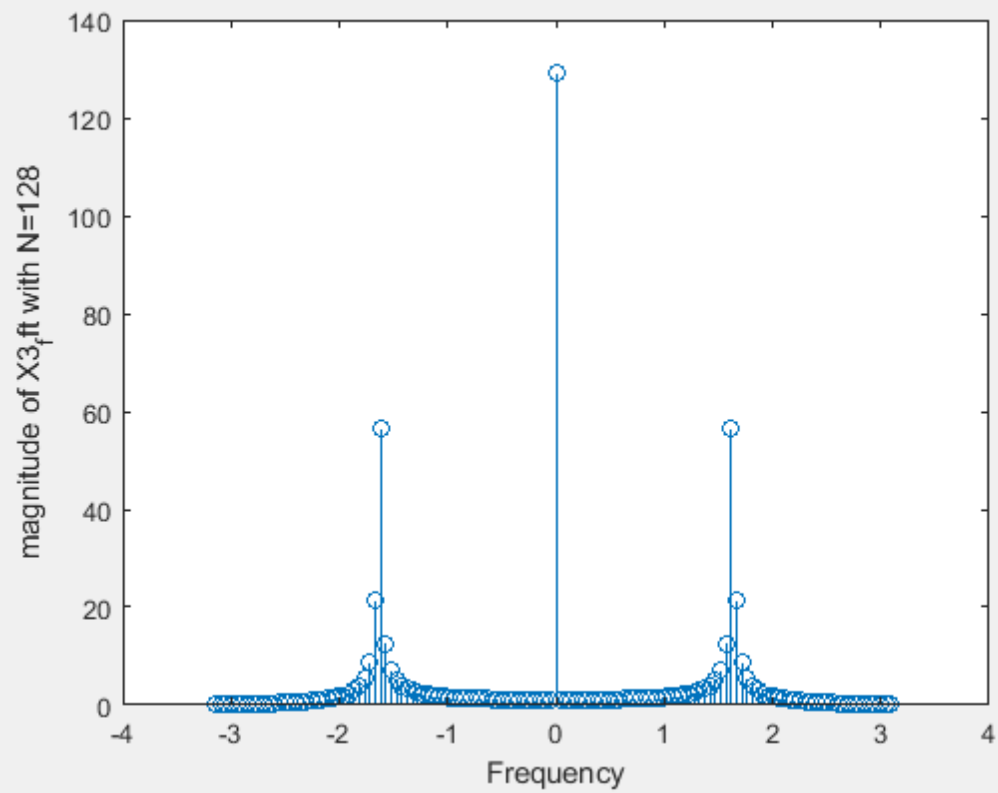
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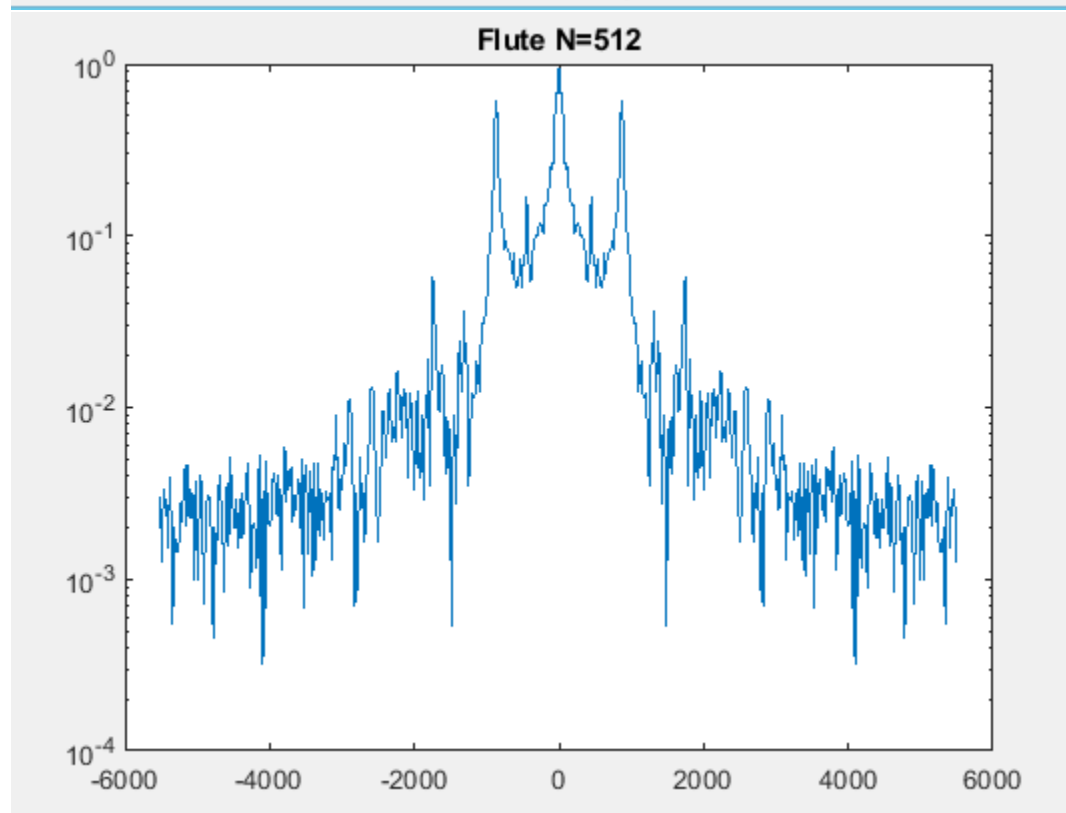
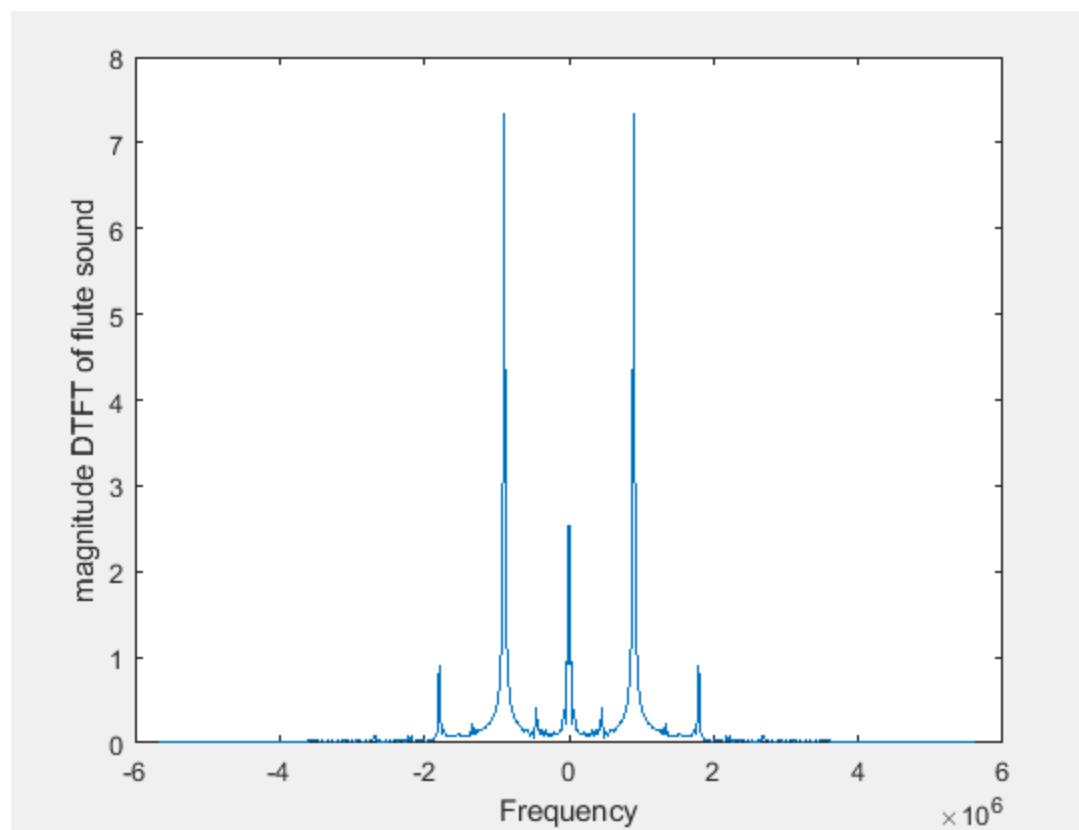
Q2: Turn in the plots for the cosine with frequency  $f=0.26$  for the 3 window lengths and explain why the DFTs look different. Include the flute DFT plot, specify what length FFT you used and why, and explain how you determined the note that is being played.

A2: The plots in frequency domain are different because the sampling frequency ( $N$ ) is different.

I used a  $N = 512$  to cover all significant frequency content. According to the plot, the highest peaks that are not at  $w=0$  are at  $\pm 861.3$  Hz, which is close to A5 (880Hz). So the note in the flute-short.wav is close to note A.







### 3 ASSIGNMENT III

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Q3 Turn in plots of the magnitude and plots for (a)-(c). State what type of signals each corresponds to (low pass, high pass, etc.) and what the (normalized) cut-off frequencies are for each. Explain why the DTFTs do not have a flat frequency response in the passband.

A3:  $x_1$  is a low-pass filter.  $x_2$  is a high-pass filter.  $x_3$  is a band-pass filter.

Cut-off frequencies:  $x_1$  (0.47),  $x_2$  (2.7),  $x_3$  (0.79 – 1.72)

Since  $x_1[t]$ ,  $x_2[t]$  and  $x_3[t]$  are defined in the range (0, 255), which doesn't cover the whole positive  $x$ -axis to  $+\infty$ , the Fourier transforms of the signals are not perfect boxes.

