



EE2703: Applied Programming Lab

Assignment 8

Discrete Fourier Transform

Arjun Menon Vadakkeveedu
EE18B104
Electrical Engineering, IIT Madras
April 20, 2020

Contents

1	Introduction	3
2	Approach	4
2.1	Important blocks used in the program	4
3	FFT Analysis	6
3.1	$\sin(5t)$	6
3.2	Amplitude Modulation	6
3.3	$\sin^3(t)$ and $\cos^3(t)$	7
3.4	Frequency Modulation	8
3.5	Gaussian	9

1 Introduction

This program uses numpy's fft module to compute the Discrete Fourier Transform of some common signals (sinusoids, AM signal, FM signal and Gaussian). The Discrete Fourier Transform of a discrete time sequence $x[n]$ is defined as:

$$X[k] = \sum_{n=0}^{N-1} x[n] e^{-jk \frac{2\pi}{N} n}$$

and the inverse DFT is obtained by:

$$x[n] = \sum_{k=0}^{N-1} X[k] e^{jk \frac{2\pi}{N} n}$$

The DFT can be interpreted as a sampling of the Discrete Time Fourier Transform (DTFT). As a result, the corresponding time domain sequence is periodic with period N .

Time and Frequency Bins: The time vector is sampled for a total of c cycles (each cycle is of duration 2π) and N samples are recorded. This means that the time vector is (defined as a non-causal signal in time) $\text{linspace}(-c\pi, c\pi, N+1)[:1]$

Similarly, the frequency vector has length N and corresponds to $[-\pi, \pi)$ in the Discrete Frequency domain and $[-F_s/2, F_s/2)$ in the Continuous Frequency Domain- where F_s is the sampling frequency.

Mapping the DFT to the Continuous Time Fourier Transform: While the DFT can accurately determine the spectrum of bandlimited periodic continuous time signals sampled at a rate greater than the Nyquist rate, the DFT of a Gaussian will have errors due to its infinite band. Hence, by taking sufficiently large values of N and F_s , the CTFT of a Gaussian may be approximated using the DFT:

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

The integral may be approximated to a finite sum as (for sufficiently large N):

$$X(j\Omega) \approx \sum_{n=-N/2}^{N/2-1} x(n\Delta t) e^{-j(\Omega\Delta t)n} (\Delta t)$$

$$= (\Delta t)X[k]$$

$$= N(\Delta t) * \text{Spectrum}(x) = 2\pi c * \text{Spectrum}(x)$$

2 Approach

The DFT is computed using the FFT algorithm, implemented using `numpy.fft` module. To centre the fft vector about the 0-frequency bin, the vector is rearranged using the `fftshift` function.

2.1 Important blocks used in the program

1: Generating FFT

```
def gen_fft(f, num_cycles, sample_freq, ii_cutoff):
    N = num_cycles*sample_freq
    x = np.linspace(-1*pi, pi, N+1)*num_cycles; x = x[:-1]
    y = f(x)
    y = ifftshift(y)
    Y = fftshift(fft(y))/N
    w = np.linspace(-0.5, 0.5, N+1)*sample_freq; w = w[:-1]
    ii = np.where(abs(Y)>ii_cutoff)
    return Y, w, ii
```

2: Plotting DFT characteristics:

```
def gen_plots(Y, w, ii, x_range, title, plot_gauss = False):
    fig = plt.figure()
    plt.subplot(2,1,1)
    plt.xlim(x_range)
    plt.plot(w, abs(Y), lw = 2)
    if plot_gauss == True:
        plt.plot(w, gauss_f(w))
    plt.ylabel(r"|Y|", size = 16)
    plt.grid(True)
    plt.title(title)
    plt.subplot(2,1,2)
    plt.xlim(x_range)
    plt.plot(w[ii], (np.angle(Y[ii])), 'ro', lw = 2)
```

```

plt.ylabel(r"Phase of Y")
plt.grid(True)
plt.tight_layout()
plt.show()
if plot_gauss == True:
    plt.xlim(x_range)
    plt.plot(w, np.real(Y - gauss_f(w)))
    plt.title("Error of DFT with true Fourier Transform")
    plt.show()
return 0

```

3: Input Format:

```

python3 EE18B104_Assign8.py --h
usage: EE18B104_Assign8.py [-h] [--fn_choice FN_CHOICE]

```

optional arguments:

```

-h, --help            show this help message and exit
--fn_choice FN_CHOICE

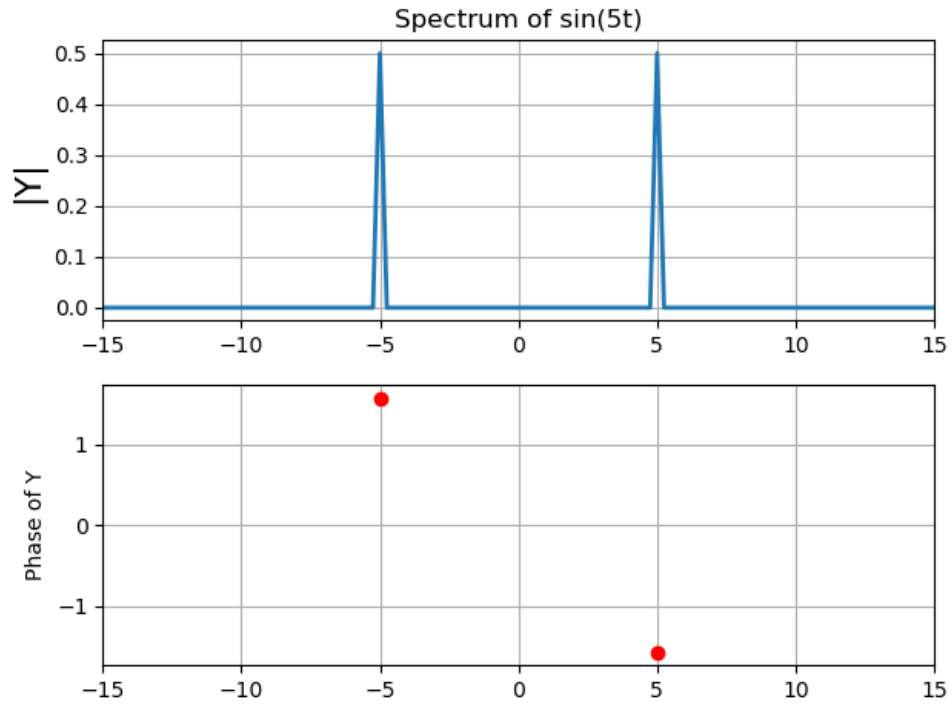
```

Choose time domain function:

- 1: $\sin(5t)$
 - 2: $(1+0.1\cos(t))\cos(10t)$ (AMPLITUDE MODULATION)
 - 3: $\sin^3(t)$, $\cos^3(t)$
 - 4: $\cos(20t + 5\cos(t))$ (FREQUENCY MODULATION)
 - 5: $\exp(-(t^2)/2)$ (GAUSSIAN)
-

3 FFT Analysis

3.1 $\sin(5t)$



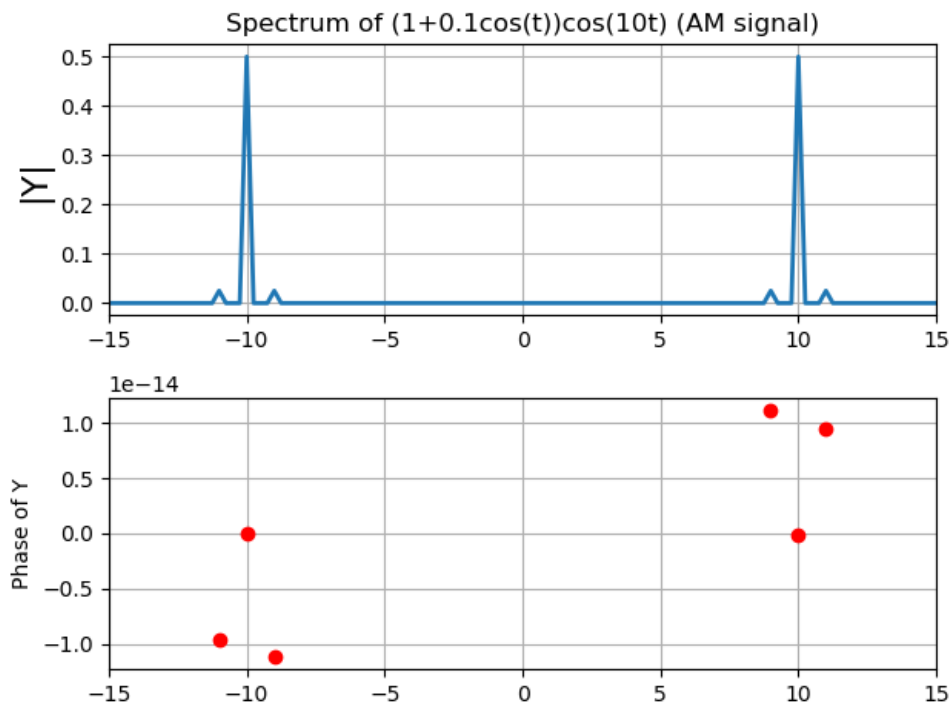
As can be seen in the figure, $\sin(5t)$ comprises of two deltas at -5 and +5 with phase $+\pi/2$ and $-\pi/2$ respectively

3.2 Amplitude Modulation

$$y(t) = (1 + 0.1\cos(t))\cos(10t) = \cos(10t) + 0.1 * \cos(t) * \cos(10t)$$

$$y(t) = \cos(10t) + 0.05 * (\cos(11t) + \cos(9t))$$

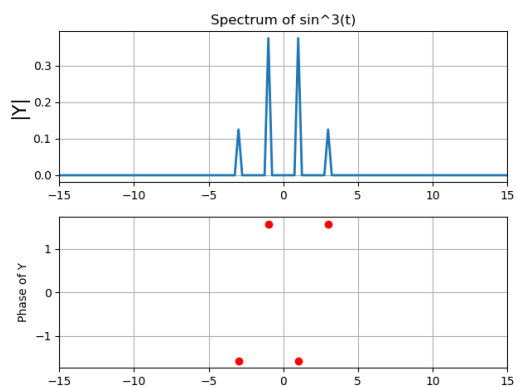
Thus, the AM signal is expected to have frequency components at ± 9 , ± 10 , ± 11 rad/s.



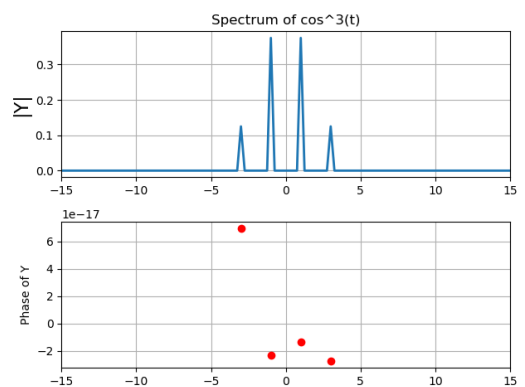
Since the signal is composed of cosines, the phase is 0 (with some finite errors due to the implementation of fft)

3.3 $\sin^3(t)$ and $\cos^3(t)$

$$\sin^3(t) = 0.75\sin(t) - 0.25\sin(3t); \cos^3(t) = 0.75\cos(t) + 0.25\cos(3t)$$



(a) $\sin^3(t)$



(b) $\cos^3(t)$

3.4 Frequency Modulation

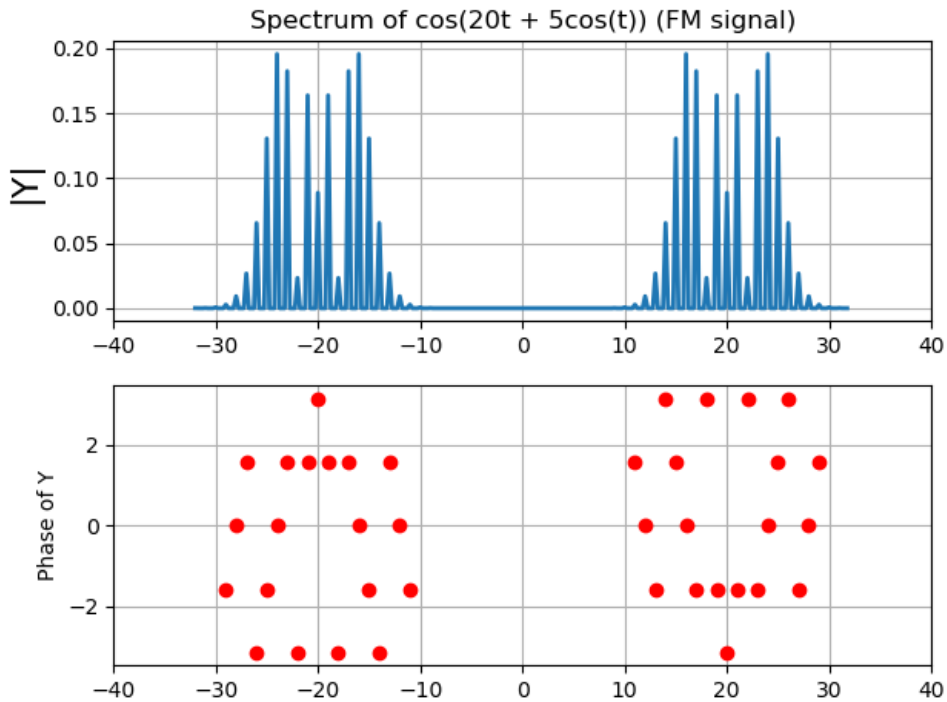
$$\cos(20t + 5\cos(t)) = \cos(20t)\cos(5\cos(t)) - \sin(20t)\sin(5\cos(t)) = \Re(e^{j20t}e^{j5\cos(t)})$$

The Fourier Series coefficients of $(e^{j5\cos(t)})$ are given by the Jacobi-Anger expansion:

$$e^{j5\cos(t)} = \sum_{n=-\infty}^{+\infty} (j^n) J_n(5) e^{jnt}$$

where $J_n(5)$ is the n-th order Bessel function of the first kind evaluated at 5. Since the DFT of e^{j20t} are impulses, and since the DFT of the signal is the convolution of the DFTs of the Bessel impulse train with impulses centred at ± 20 .

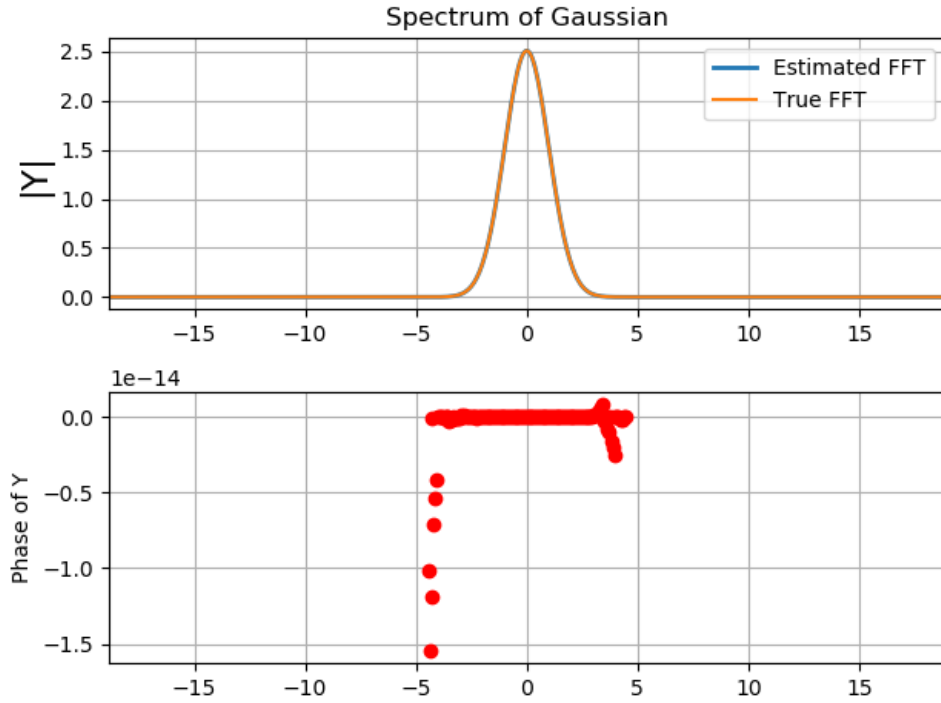
This is clearly visible from the magnitude plot of the DFT. As for the phase, note that $J_n(5)$ varies as $(-1)^n$. About $\omega = 20$, the phase of the DFT varies as $(-j)^{\text{mod}(n-20,20)}$ (and similarly for $\omega = -20$)



3.5 Gaussian

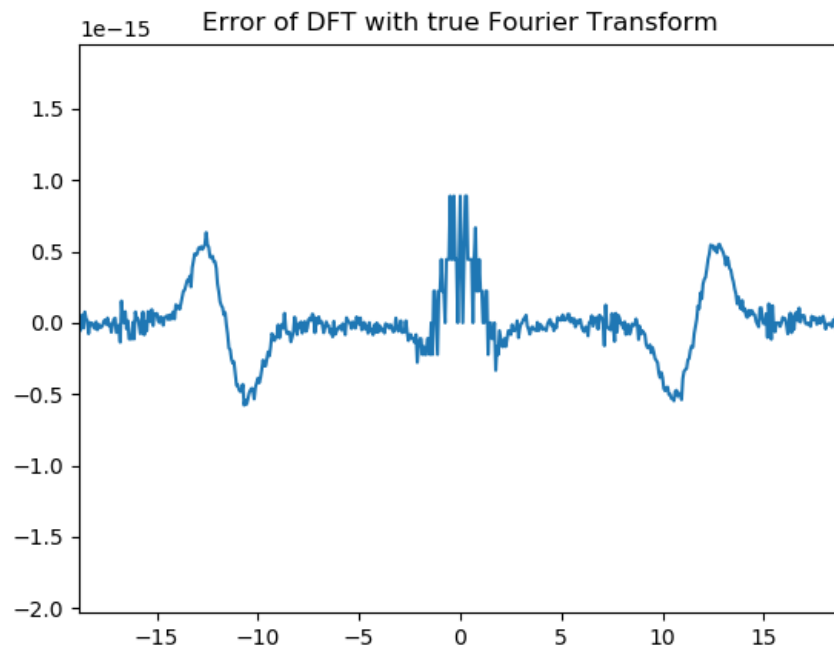
As explained above, the CTFT of the Gaussian signal is estimated using the DFT. The following plot was obtained for $F_s = 128\text{Hz}$, for a total time duration of 32π ($N = 2048$)

$$x(t) = e^{-t^2/2}$$
$$X(j\Omega) = \sqrt{2\pi}e^{-\Omega^2/2}$$



The error between the estimated FFT and the true FFT is quite low. As the FFT of Gaussian is real, the phase is 0

The error between the estimated and true FFT is plotted as a function of ω :



Error between the Estimated and True FFTs of Gaussian- note that in addition to this, the estimated FFT also has an imaginary part of the order of $1e-16$
