

Assignment 8

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1 Introduction

This week's assignment deals with to learn about DFT and how it is implemented in python using Numpy's FFT module. FFT is an implementation of the DFT. We also attempt to approximate the continuous time fourier transform of a gaussian by windowing and sampling in time domain, and then taking the DFT. For Understanding the plots are plotted.

2 FFT and IFFT

We find the Fourier transform and invert it back to the time domain for a random signal, find maximum error to test the reconstruction

The maximum error obtained from the below code = $2.7781647035173285 \times 10^{-16}$

```
x=rand(100)
X=fft(x)
y=ifft(X)
c=[x,y]
print(abs(x-y).max())
```

3 Spectrum of $\sin(5t)$

The solution for this is already a part of the assignment. As expected the phase for some values near the peaks is non zero. To fix this we sample the input signal at an appropriate frequency. We also shift the phase plot so that it goes from $-\pi$ to π . To do this we write a helper function

```
x=linspace(0,2*pi,129);x=x[:-1]
y=sin(5*x)
Y=fftshift(fft(y))/128.0
w=linspace(-64,63,128)
```

```

figure(0)
subplot(2,1,1)
plot(w,abs(Y),lw=2)
xlim([-10,10])
ylabel(r"$|Y|$",size=16)
title(r"Spectrum of \sin(5t)")
grid(True)
subplot(2,1,2)
plot(w,angle(Y),'ro',lw=2)
ii=where(abs(Y)>1e-3)
plot(w[ii],angle(Y[ii]),'go',lw=2)
xlim([-10,10])
ylabel(r"Phase of Y",size=16)
xlabel(r"$k$",size=16)
grid(True)
show()

```

As expected we get 2 peaks at +5 and -5 with height 0.5. The phases of the peaks at $\frac{\pi}{2}$ and $-\frac{\pi}{2}$ are also expected based on the expansion of a sine wave ie:

$$\sin(5t) = 0.5\left(\frac{e^{5t}}{j} - \frac{e^{-5t}}{j}\right) \quad (1)$$

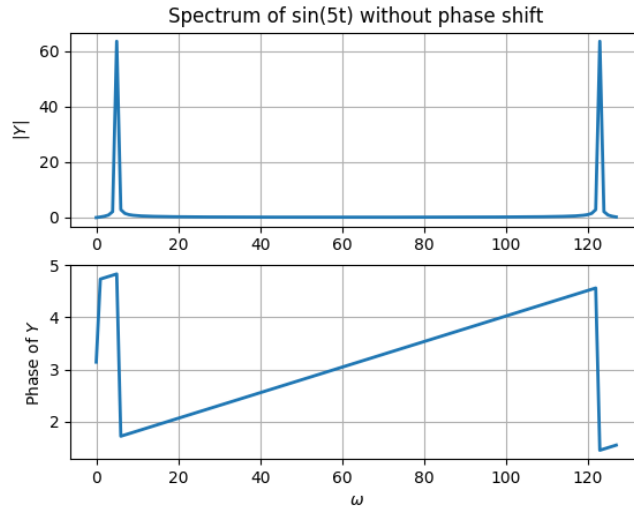


Figure 1: Spectrum of $\sin(5t)$ without Phase wrapping

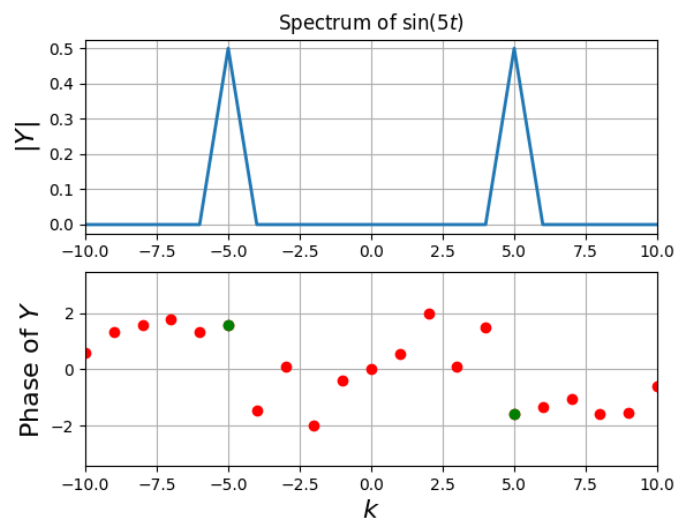


Figure 2: Spectrum of $\sin(5t)$ after phase wrapping

4 Amplitude Modulation

Consider the signal:

$$f(t) = (1 + 0.1 \cos(t)) \cos(10t) \quad (2)$$

We expect a shifted set of spikes, with a main impulse and two side impulses on each side. This is because,

$$0.1 \cos(10t) \cos(t) = 0.05(\cos 11t + \cos 9t) = 0.025(e^{11tj} + e^{9tj} + e^{11tj} + e^{9tj}) \quad (3)$$

In order to see even the side peaks, the frequency resolution has to be improved. We can do so by keeping the number of samples constant and increasing the range in the time domain. The following spectrum is obtained. The python code for this part is shown below:-

```
t1=linspace(-4*pi,4*pi,513);t1=t1[: -1]
y1=(1+0.1*cos(t1))*cos(10*t1)
Y1=fftshift(fft(y1))/512.0
w1=linspace(-64,64,513);w1=w1[: -1]
figure(1)
subplot(2,1,1)
plot(w1,abs(Y1),lw=2)
xlim([-15,15])
ylabel(r"$|Y|$",size=16)
title(r"Spectrum of $\left(1+0.1\cos\left(t\right)\right)\cos\left(10t\right)$")
grid(True)
subplot(2,1,2)
plot(w1,angle(Y1),'ro',lw=2)
xlim([-15,15])
ylabel(r"Phase of $Y$",size=16)
xlabel(r"$\omega$",size=16)
grid(True)
show()
```

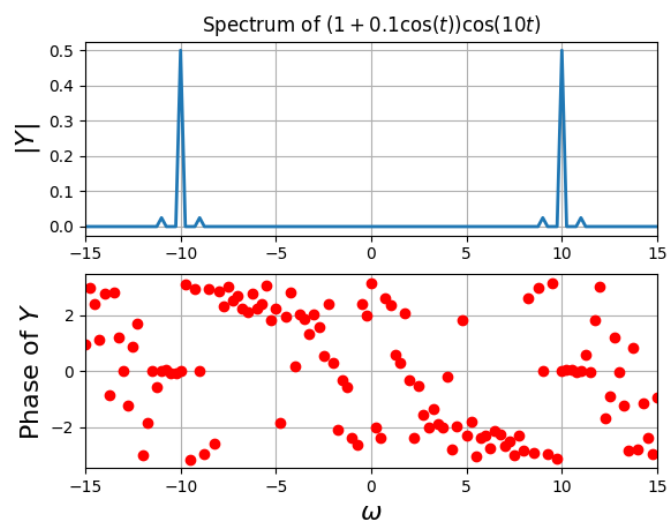


Figure 3: Spectrum of $f(t) = (1 + 0.1 \cos(t)) \cos(10t)$ with a higher number of samples

5 Spectrum of $\sin^3(t)$

This signal can be expressed as a sum of sine waves using this identity:

$$\sin^3(t) = \frac{3}{4} \sin(t) - \frac{1}{4} \sin(3t)$$

We expect 2 peaks at frequencies 1 and 3, and phases similar to that expected from a sum of sinusoids. The python code for this part:-

```
t3=linspace(-4*pi,4*pi,513);t3=t3[: -1]
y3=(sin(t3))**3
Y3=fftshift(fft(y3))/512.0
w3=linspace(-64,64,513);w3=w3[: -1]
figure(3)
subplot(2,1,1)
plot(w3,abs(Y3),lw=2)
xlim([-15,15])
ylabel(r"$|Y|$" , size=16)
title(r"Spectrum of $\sin^3(t)$")
grid(True)
subplot(2,1,2)
plot(w3,angle(Y3), 'ro', lw=2)
xlim([-15,15])
ylabel(r"Phase of $Y$" , size=16)
xlabel(r"$\omega$", size=16)
grid(True)
show()
```

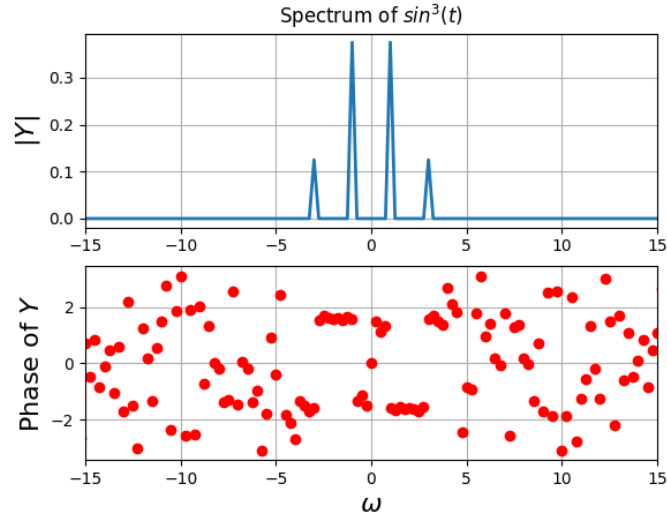


Figure 4: Spectrum of $f(t) = \sin^3(t)$

6 Spectrum of $\cos^3(t)$

This signal can be expressed as a sum of cosine waves using this identity:

$$\sin^3(t) = \frac{3}{4} \cos(t) + \frac{1}{4} \cos(3t)$$

We expect 2 peaks at frequencies 1 and 3, and phases similar to that expected at the peaks. The python code for this part is given below:-

```
t2=linspace(-4*pi,4*pi,513);t2=t2[: -1]
y2=(cos(t2))**3
Y2=fftshift(fft(y2))/512.0
w2=linspace(-64,64,513);w2=w2[: -1]
figure(2)
subplot(2,1,1)
plot(w2,abs(Y2),lw=2)
xlim([-15,15])
ylabel(r"$|Y|$" , size=16)
title(r"Spectrum of $\cos^3(t)$")
grid(True)
subplot(2,1,2)
plot(w2,angle(Y2), 'ro', lw=2)
xlim([-15,15])
ylabel(r"Phase of $Y$" , size=16)
xlabel(r"$\omega$", size=16)
grid(True)
show()
```

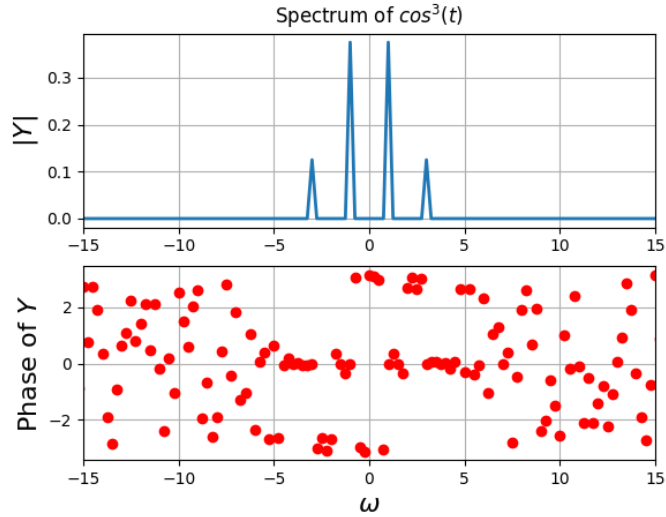


Figure 5: Spectrum of $f(t) = \cos^3(t)$

7 Spectrum of Frequency Modulated Wave

Consider the signal:

$$f(t) = \cos(20t + 5 \cos(t)) \quad (4)$$

Using the same helper function as before, we get the following output:
The python code for this part is:-

```
t4=linspace(-4*pi,4*pi,513);t4=t4[: -1]
y4=cos(20*t4 + 5*cos(t4))
Y4=fftshift(fft(y4))/512.0
w4=linspace(-64,64,513);w4=w4[: -1]
figure(1)
subplot(2,1,1)
plot(w4,abs(Y4),lw=2)
xlim([-30,30])
ylabel(r"$|Y|$" , size=16)
title(r"Spectrum of cos(20t + 5cos(t))")
grid(True)
subplot(2,1,2)
ii=where(abs(Y4)>1e-3)
plot(w4[ii],angle(Y4[ii]),'go',lw=2)
xlim([-30,30])
ylabel(r"Phase of Y" , size=16)
xlabel(r"$\omega$" , size=16)
grid(True)
show()
```

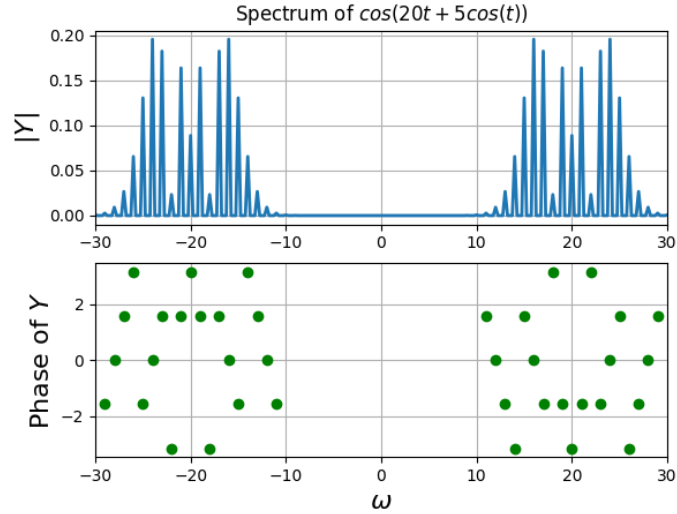


Figure 6: Spectrum of $f(t) = (1 + 0.1 \cos(t)) \cos(10t)$

The number of peaks has clearly increased. The energy in the side bands is comparable to that of the main signal.

8 Continuous time Fourier Transform of a Gaussian

The Fourier transform of a signal $x(t)$ is defined as follows:

$$X(\omega) = \frac{1}{2\pi} \int_{-\infty}^{\infty} x(t) e^{-j\omega t} dt \quad (5)$$

We can approximate this by the Fourier transform of the windowed version of the signal $x(t)$, with a sufficiently large window as Gaussian curves tend to 0 for large values of t . Let the window be of size T . We get:

$$X(\omega) \approx \frac{1}{2\pi} \int_{-T/2}^{T/2} x(t) e^{-j\omega t} dt \quad (6)$$

On writing the integral as a Riemann sum with a small time step $\Delta t = \frac{T}{N}$, We get:

$$X(\omega) \approx \frac{\Delta t}{2\pi} \sum_{n=-\frac{N}{2}}^{\frac{N}{2}-1} x(n\Delta t) e^{-j\omega n\Delta t} \quad (7)$$

Now, we sample our spectrum with a sampling period in the frequency domain of $\Delta\omega = \frac{2\pi}{T}$, which makes our continuous time signal periodic with period equal to the window size T . Our transform then becomes:

$$X(k\Delta\omega) \approx \frac{\Delta t}{2\pi} \sum_{n=-\frac{N}{2}}^{\frac{N}{2}-1} x(n\Delta t) e^{-j\frac{2\pi}{N} kn} \quad (8)$$

This form is similar to a DFT(for a finite window size). Therefore:

$$X(k\Delta\omega) \approx \frac{\Delta t}{2\pi} DFT\{x(n\Delta t)\} \quad (9)$$

We made a few approximations by using a finite window size and by using the Riemann approximation

We can improve these approximations by making the window size T larger, and by decreasing the time domain sampling period or increasing the number of samples N . We find the appropriate values for these iterative keeping the sampling frequency constant.

The expression for the Gaussian is :

$$x(t) = e^{-\frac{t^2}{2}} \quad (10)$$

The CTFT is given by:

$$X(j\omega) = \frac{1}{\sqrt{2\pi}} e^{\frac{-\omega^2}{2}} \quad (11)$$

```

def gauss(x):
    return exp(-0.5*x**2)

def expectedgauss(w):
    return 1/sqrt(2*pi) * exp(-w**2/2)

def estdft(tolerance=1e-6,samples=128,func = gauss,expectedfn = expectedgauss,wlim = 5):
    T = 8*pi
    N = samples
    Yold=0
    err=tolerance+1
    iters = 0
    #iterative loop to find window size
    while err>tolerance:
        x=linspace(-T/2,T/2,N+1)[: -1]
        w = linspace(-N*pi/T,N*pi/T,N+1)[: -1]
        y = gauss(x)
        Y=fftshift(fft(iffshift(y)))*T/(2*pi*N)
        err = sum(abs(Y[:,2] - Yold))
        Yold = Y
        iters+=1
        T*=2
        N*=2

    true_error = sum(abs(Y-expectedfn(w)))
    print("True_error: ",true_error)
    print(" samples=",str(N)+" time_period=",pi*str(T/pi))

    mag = abs(Y)
    phi = angle(Y)
    phi[where(mag<tolerance)]=0

    figure()
    subplot(2,1,1)
    plot(w,abs(Y),lw=2)
    xlim([-wlim,wlim])
    ylabel('Magnitude',size=16)
    title("Estimate_fft_of_gaussian")
    grid(True)
    subplot(2,1,2)
    plot(w,angle(Y),'ro',lw=2)
    ii=where(abs(Y)>1e-3)
    plot(w[ii],angle(Y[ii]),'go',lw=2)
    xlim([-wlim,wlim])
    ylabel("Phase",size=16)
    xlabel("w",size=16)
    grid(True)
    show()

    Y_ = expectedfn(w)

    mag = abs(Y_)

```

```

phi = angle(Y_)
phi[where(mag<tolerance)]=0

figure()
subplot(2,1,1)
plot(w,abs(Y),lw=2)
xlim([-wlim,wlim])
ylabel('Magnitude',size=16)
title("True\_fft\_of\_gaussian")
grid(True)
subplot(2,1,2)
plot(w,angle(Y),'ro',lw=2)
ii=where(abs(Y)>1e-3)
plot(w[ii],angle(Y[ii]),'go',lw=2)
xlim([-wlim,wlim])
ylabel("Phase",size=16)
xlabel("w",size=16)
grid(True)
show()

return

estdft()

```

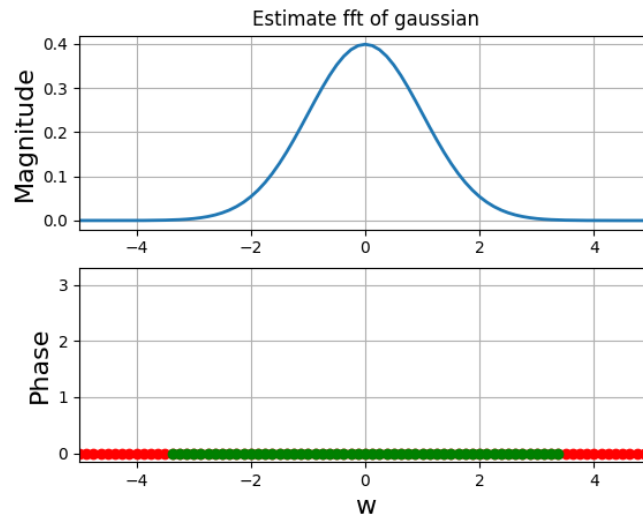


Figure 7: estimated CTFT of Gaussian

The summation of the absolute values of the error, Final number of samples and final window size are given as:
True error: 1.472553842671434e-14
samples = 512 time period = $\pi \cdot 32.0$

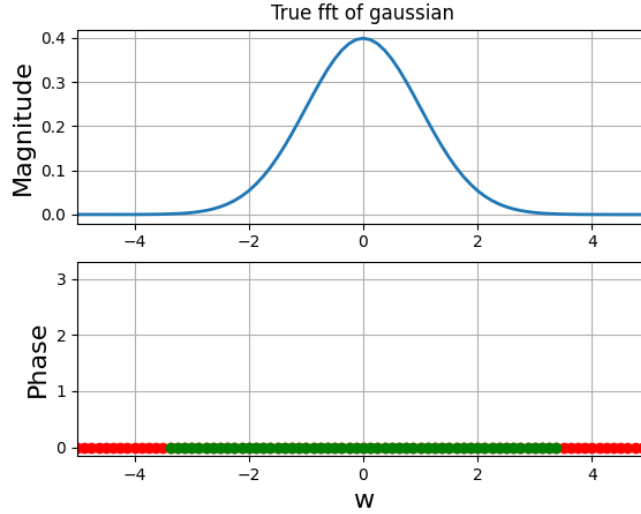


Figure 8: Expected CTFT of Gaussian

From the above pairs of plots, it is clear that with a sufficiently large window size and sampling rate, the DFT approximates the CTFT of the gaussian. This is because the magnitude of the gaussian quickly approaches 0 for large values of time. This means that there is lesser frequency domain aliasing due to windowing. Windowing in time is equivalent to convolution with a sinc in frequency domain. A large enough window means that the sinc is tall and thin. This tall and thin sinc is approximately equivalent to a delta function for a sufficiently large window. This means that convolution with this sinc does not change the spectrum much. Sampling after windowing is done so that the DFT can be calculated using the Fast Fourier Transform. This is then a sampled version of the DTFT of the sampled time domain signal. With sufficiently large sampling rates, this approximates the CTFT of the original time domain signal. This process is done on the gaussian and the results are in agreement with what is expected.

9 Conclusion

The `fft` library in python provides a useful toolkit for analysis of DFT of signals. The Discrete Fourier Transforms of sinusoids, amplitude modulate signals, frequency modulated signals were analysed. In the case of pure sinusoids, the DFT contained impulses at the sinusoid frequencies. The amplitude modulated wave had a frequency spectrum with impulses at the carrier and the side band frequencies. The frequency modulated wave, having an infinite number of side band frequencies, gave rise a DFT with non zero values for a broader range of frequencies. The DFT of a gaussian is also a gaussian and the spectrum was found to sharpen for higher sampling rates, while broaden for greater time ranges.