# **American International University- Bangladesh Department of Computer Engineering**

COE 3201: Data Communication Laboratory

<u>Title:</u> Study of signal frequency, spectrum, bandwidth, bit rate, quantization using MATLAB

## **Abstract:**

This experiment is designed to-

- 1.To understand the use of MATLAB for solving communication engineering problems.
- 2.To develop understanding of MATLAB environment, commands and syntax.

### **Introduction:**

I. **Frequency:** The frequency of a wave describes how many waves go past a certain point in one second. Frequency is measured in Hertz (usually abbreviated Hz), and can be calculated using the formula:

$$V = f\lambda$$

where V is the velocity of the wave (in  $ms^{-1}$ ), f is the frequency of the wave (in Hz), and  $\lambda$  (the Greek letter lambda) is the wavelength of the wave (distance from one peak / trough to the next, in m). Frequency is the rate of change with respect to time. Change in a short span of time means high frequency. Change over a long span of time means low frequency.

II. **Spectrum:** Usually we represent signals in time domain. But signals can be represented in frequency domain as well. When signals are represented in frequency domain they are called spectrum.

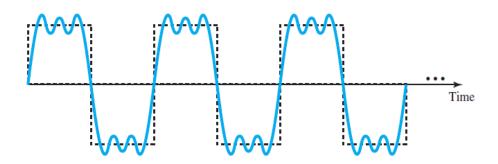


Fig: A composite periodic signal

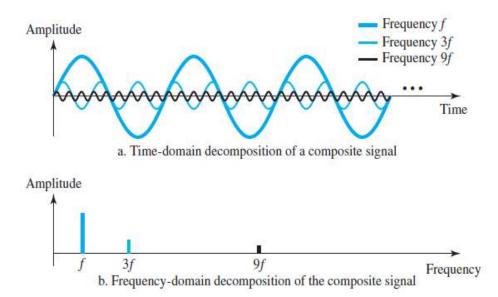
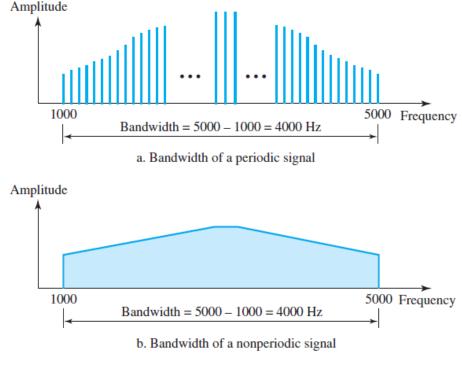


Fig: Decomposition of a composite periodic signal in the time and frequency domains

III. **Bandwidth:** Bandwidth is the range of frequency a signal contains in it. If a composite signal is made up of multiple sinusoids of 100, 250, 300, and 400 Hz. Then its bandwidth is the difference of the highest and lowest frequency components. So here the bandwidth of the signal is (400-100) = 300 Hz.

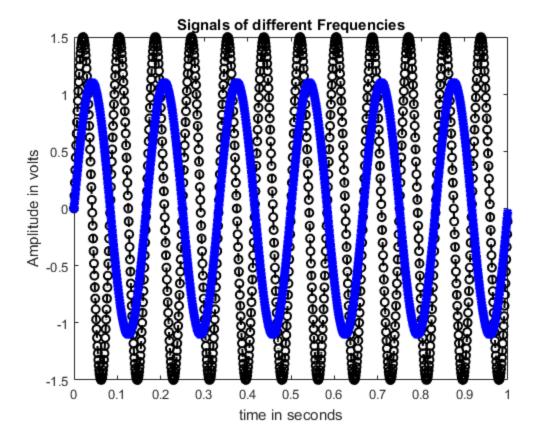


IV. **Bit rate:** Bit Rate, sometimes also written as bitrate, indicates the amount of data that is processed or delivered over a certain time span. The data is measured in bits while the time is measured in seconds, resulting in a bit/s or bps (bits by seconds) unit. When dealing with a high Bit Rate, the unit can also be expressed in kilobit (kbps), megabit (mbps), gigabit (gbps), and so on.

When talking about the Bit Rate of audio or video files, it often affects the quality of the audio data in both cases. As a rule of thumb, the higher the Bit Rate, the better the audio quality. For example, an MP3 file with a Bit Rate of 192 kbps will have a considerably better quality than a file that has a Bit Rate of only 32 kbps. Furthermore, the Bit Rate affects the image quality of a video file.

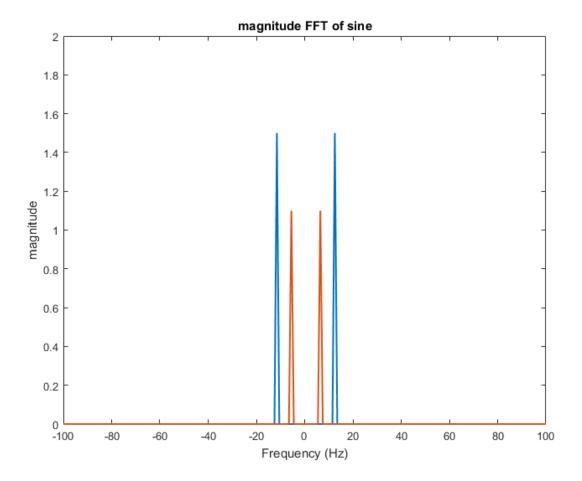
Generating a sinusoidal signal with different frequencies:

```
fs = 1000; % Sampling frequency
t = 0:1/fs:1-1/fs; % Time duration
f1 = 12; % Frequency of first signal
f2 = 6; % Frequency of second signal
A1 = 1.5; % Amplitude of first signal
A2 = 1.1; % Amplitude of second signal
x1 = A1*sin(2*pi*f1*t); % First Signal
x2 = A2*sin(2*pi*f2*t); % Second Signal
%Plotting both signals in time domain
plot(t,x1,'k--o','LineWidth',1.5)
hold on
plot(t, x2, 'b-*', 'LineWidth', 2)
hold off
xlabel('time in seconds')
vlabel('Amplitude in volts')
title('Signals of different Frequencies')
```



These signals can be represented in frequency domain as well:

```
%Take fourier transform
fx1 = fft(x1);
fx2 = fft(x2);
%apply fftshift to put it in the form we are used to (see
documentation)
fx1 = fftshift(fx1)/(fs/2);
fx2 = fftshift(fx2)/(fs/2);
%Next, calculate the frequency axis, which is defined by the
sampling rate
f = fs/2*linspace(-1,1,fs);
%Since the signal is complex, we need to plot the magnitude to
get it to
%look right, so we use abs (absolute value)
figure;
plot(f, abs(fx1), f, abs(fx2), 'LineWidth', 1.5);
title ('magnitude FFT of sine');
axis([-100 100 0 2])
xlabel('Frequency (Hz)');
ylabel('magnitude');
```



Similar task can be done where we use a composite signal instead of signals x1 and x2. Suppose our composite signal is

$$cx = 1.1*sin(2*pi*2*t)+1.3*cos(2*pi*10*t)+1.5*sin(2*pi*20*t);$$

\*\*\*\*\*Show this signal in time domain and frequency domain representation.

Example of Bandwidth calculation:

```
fs = 8000; % Sampling frequency
t = 0:1/fs:1-1/fs; % Time duration
cx = 1.1*sin(2*pi*100*t) + 1.3*cos(2*pi*300*t) +
1.5*sin(2*pi*2000*t);
bandwidth = obw(cx,fs)
```

#### bandwidth =

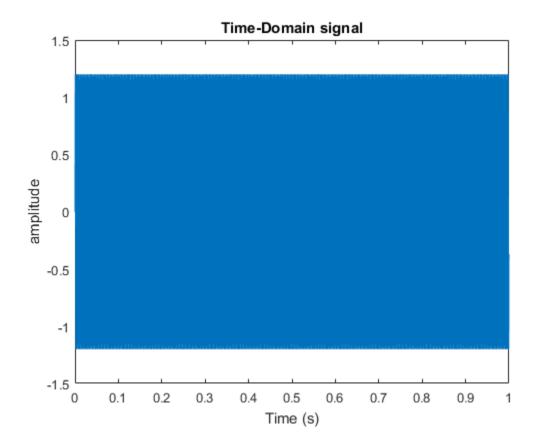
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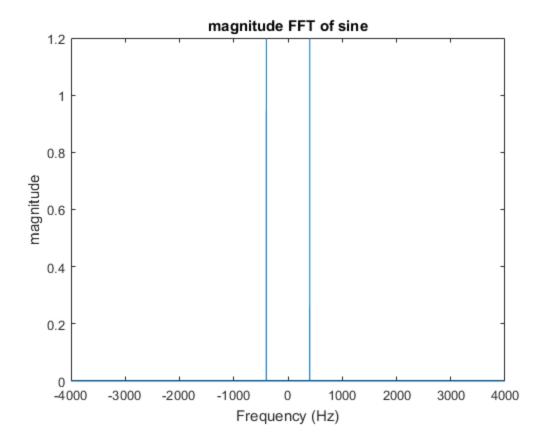
Another example of generating signals in time domain, adding noise and representing these signals in frequency domain

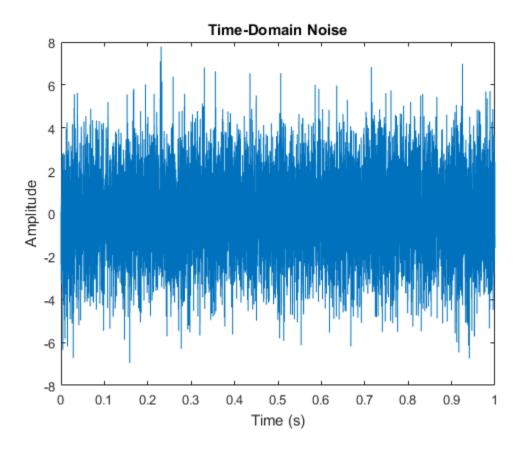
```
close all;
clc;
%Define number of samples to take
fs = 8000;
f = 400; %HZ
%Define signal
t = 0:1/fs:1-1/fs;
signal = 1.2*sin(2*pi*f*t);
%Plot to illustrate that it is a sine wave
plot(t, signal);
title('Time-Domain signal');
xlabel('Time (s)');
ylabel('amplitude');
%Take fourier transform
fftSignal = fft(signal);
%apply fftshift to put it in the form we are used to (see
documentation)
fftSignal = fftshift(fftSignal)/(fs/2); % scaling done by dividing
with (fs/2)
%Next, calculate the frequency axis, which is defined by the
sampling rate
f = fs/2*linspace(-1,1,fs);
%Since the signal is complex, we need to plot the magnitude to
get it to
%look right, so we use abs (absolute value)
figure;
plot(f, abs(fftSignal));
title ('magnitude FFT of sine');
xlabel('Frequency (Hz)');
ylabel('magnitude');
%noise
noise = 2*randn(size(signal));
figure
plot(t, noise)
xlabel('Time (s)');
ylabel('Amplitude');
title('Time-Domain Noise');
fftNoise = fft(noise);
fftNoise = fftshift(fftNoise)/(fs/2);
plot(f,abs(fftNoise))
title ('Magnitude FFT of noise');
xlabel('Frequency (Hz)');
ylabel('magnitude');
%noisy signal
noisySignal = signal + noise;
figure
```

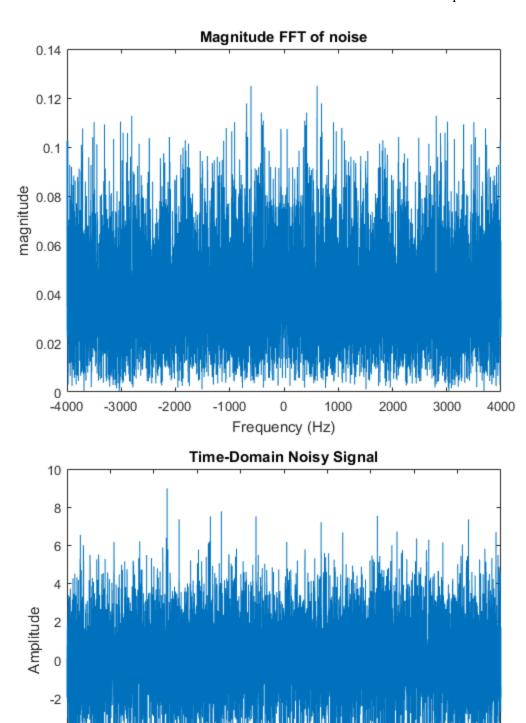
```
plot(t,noisySignal)
xlabel('Time (s)');
ylabel('Amplitude');
title('Time-Domain Noisy Signal');
fftNoisySignal = fft(noisySignal);
fftNoisySignal = fftshift(fftNoisySignal)/(fs/2);
figure
plot(f,abs(fftNoisySignal))
title('Magnitude FFT of noisy signal');
xlabel('Frequency (Hz)');
ylabel('magnitude');
```

Generated figures from this example:









0.5

Time (s)

0.4

0.6

8.0

0.9

1

0.7

0.3

-4

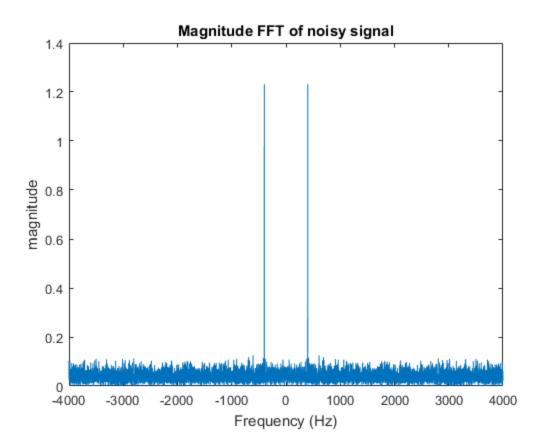
-6

-8

0

0.1

0.2



## **Performance Task for Lab Report: (your ID = AB-CDEFG-H)**

\*\*Generate a composite signal using two simple signals as,

$$x_1(t) = A_1 \cos(2\pi(C^*100)t)$$
  $x_2(t) = A_2 \cos(2\pi(F^*100)t)$   $x_3(t) = x_1(t) + x_2(t)$ 

- (a) Select the value of the amplitudes as follows: let  $A_1 = GD$  and  $A_2 = AF$ .
- (b) Make a plot of  $x_3$  over a range of t that will exhibit approximately 2 cycles. Make sure the plot starts at a negative time so that it will include t=0, and make sure that you have at least 20 samples per period of the wave.
- (c) Plot x<sub>3</sub> in frequency domain and calculate its bandwidth.
- (d) Add noise to the x3(t) signal and analysis the signal in frequency domain and show the output signal. Does the bandwidth remain same or it changes? Give your opinion. [hints: noise can be generated by rand() and make sure the matrix size remain to the x3(t) ]