## 1. Basic of Sinewave signals

I. 
$$y(t) = A \sin(2\pi F t + \phi)$$

A – Amplitude of the sine wave

F – Frequency of the sine wave

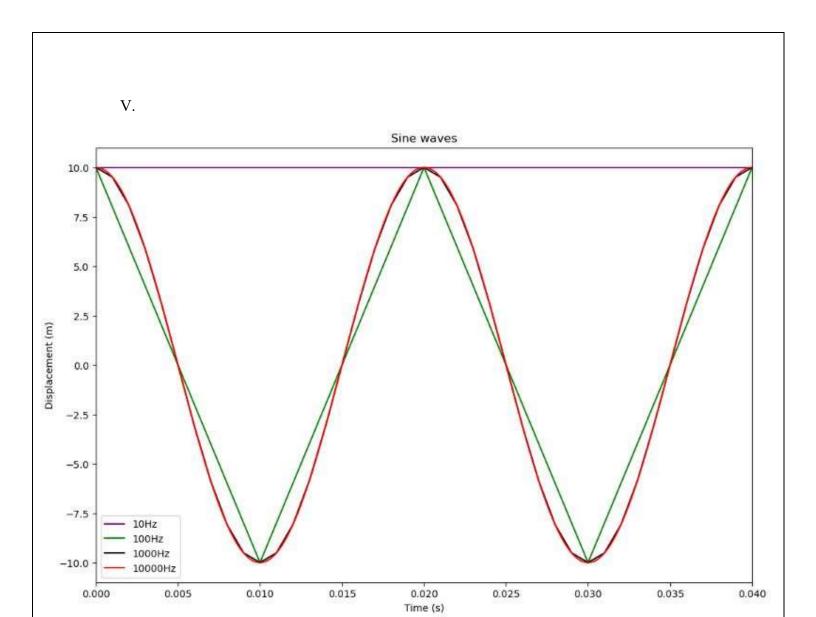
t - time

 $\emptyset$  – phase change

II. 
$$T = \frac{1}{f}$$

III. 
$$T = \frac{1}{f}$$

IV. Sampling rate is the number of samples taken from a continuous signal per second to create a digital signal. So, when the sampling rate increases it allows for more accurate capture of peak values and signal detail which can lead to better resolution and lesser distortion. It would be more accurate the stored information and the signal reconstruction from its samples. However, high sampling rate produces a large volume of data to be stored and makes necessary the use of a very fast analog-to-digital converter.



0.025

0.030

0.040

0.005

0.000

0.010

## 2. Discrete Fourier Transform (DFT)

- I. Time domain signal processing analyzes the input signal depending on the waveforms observed over a period of time. Time domain techniques emphasize the amplitude variation in a specific time. This has many applications in speech processing and heavy vehicle classification. Time is represented in the x axis and amplitude is represented in the y axis
- II. Frequency domain is an analysis of signals in reference to frequency, instead of time.
   Frequency domain explains how much of the signal exists within a given frequency band concerning a range of frequencies.
   Fourier transformation converts a time function into an integral of sine waves of various frequencies or sum, each of which symbolizes a frequency component.

III.

The main difference between Continuous Fourier Transformation (CFT) and
Discontinuous Fourier Transformation (DCFT) is that CTFT is used for continuous
time signals, while the DFT is used for discontinuous time signals.

Continuous Fourier Transformation	Discontinuous Fourier Transformation
(CFT)	(DCFT)
used for continuous time signals and	used for discontinuous time signals and
aperiodic signals	
Produces a continuous and aperiodic	Produces a periodic frequency domain
frequency domain	
Extends over an infinite frequency range	Defined by its behavior over a
	frequency range of $2\pi$

a.) How does the Fourier Transform convert a discrete signal from the time domain to the frequency domain? Provide an expression.

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

N – number of samples

n – current sample

k – current frequency

 $x_n$  – sine value of the n<sup>th</sup> sample

 $X_k$  – DFT value

b.) How does the Fourier Transform convert a continues signal from the time domain to the frequency domain? Provide an expression.

$$X_f = \int x_t e^{-j2\pi f t} dt$$

N-number of samples

f – current frequency

 $x_t$  – sine value of the n<sup>th</sup> sample

 $X_f$  – CFT value

*t*− time

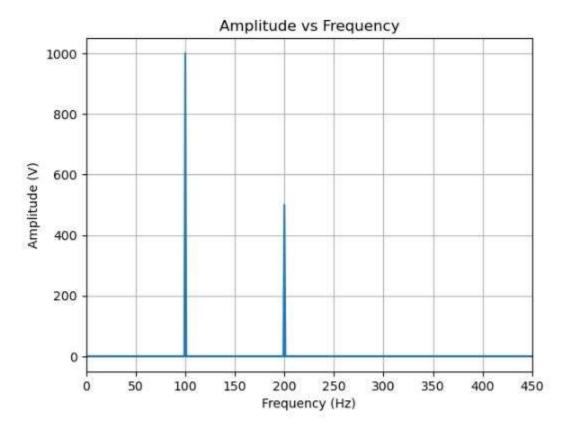
## 3. Fast Fourier Transform (FFT)

I. Fast Fourier Transform (FFT) is more efficient than Discrete Fourier Transform (DFT).

FFT uses lesser memory than DFT

Algorithm of FFT is simpler than DFT as it breaks down the DFT into small sub parts.

II.



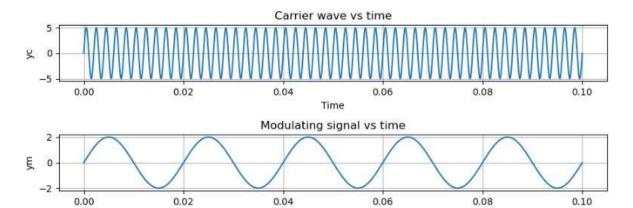
III.

a.) 
$$y_c(t) = 5\sin(2\pi \times 500 \times t)$$

b.) 
$$y_m(t) = 2\sin(2\pi \times 50 \times t)$$

c.) 
$$y(t) = (5 + 2\sin(2\pi \times 50 \times t)) \times 5\sin(2\pi \times 500 \times t)$$

d.)



Time

Magnitude of the signal vs time

0.04 0.06 0.08 0.10

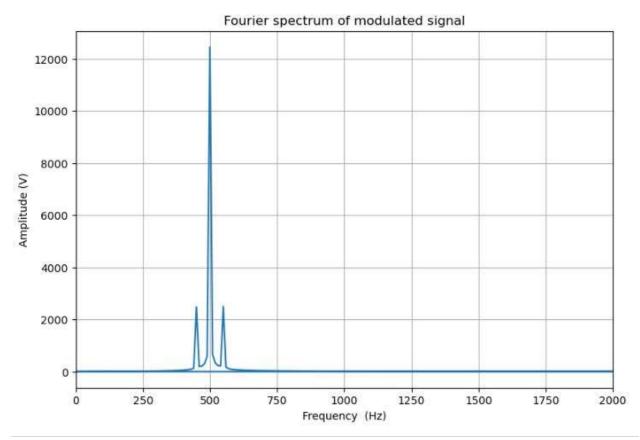
e.)

-25

0.00

0.02

i.)



ii.
Lowest frequency = 0.0 Hz

iii.
Highest frequency = 499.5 Hz

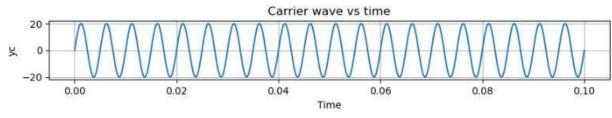
iv.
Bandwidth = 499.5 Hz

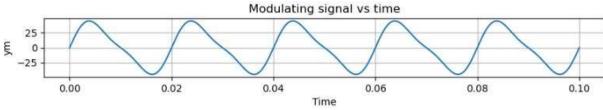
IV.)

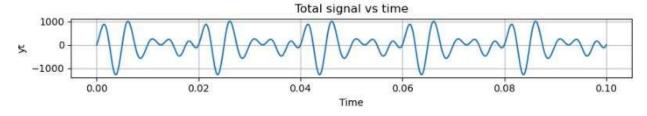
a.)

 $y(t) = 20\sin(400\pi t) (10\sin(200\pi t) + 40\sin(100\pi t))$ 

b.)

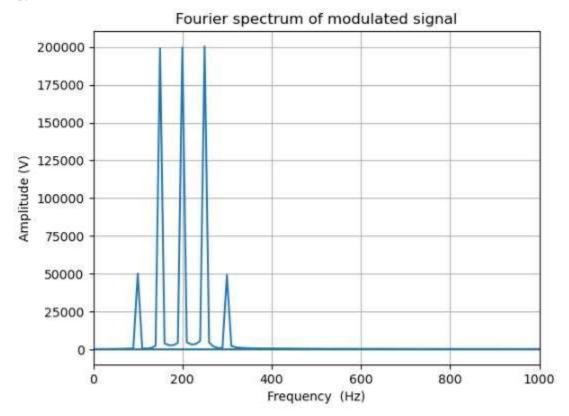






c.)

i.



ii. Lowest frequency = 0.0 Hz

iii. Highest frequency = 249.75 Hz

iv.

Bandwidth: 249.75 Hz