

Digital Signal Processing

# Discrete Fourier Transform

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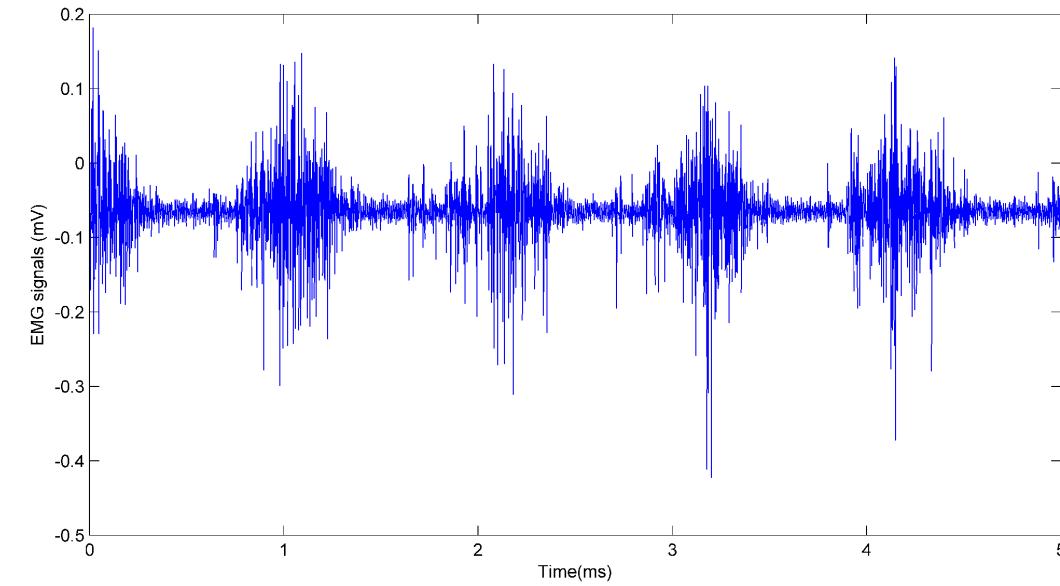
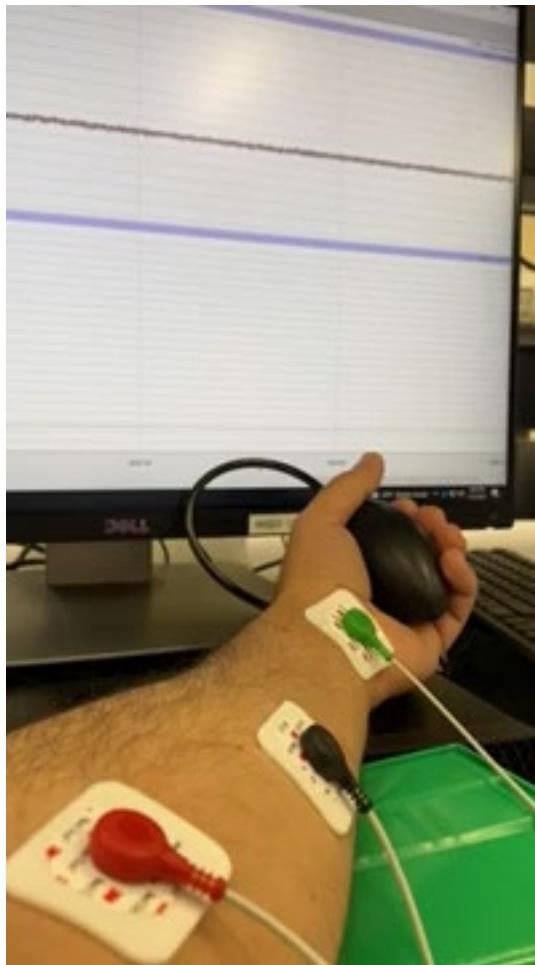
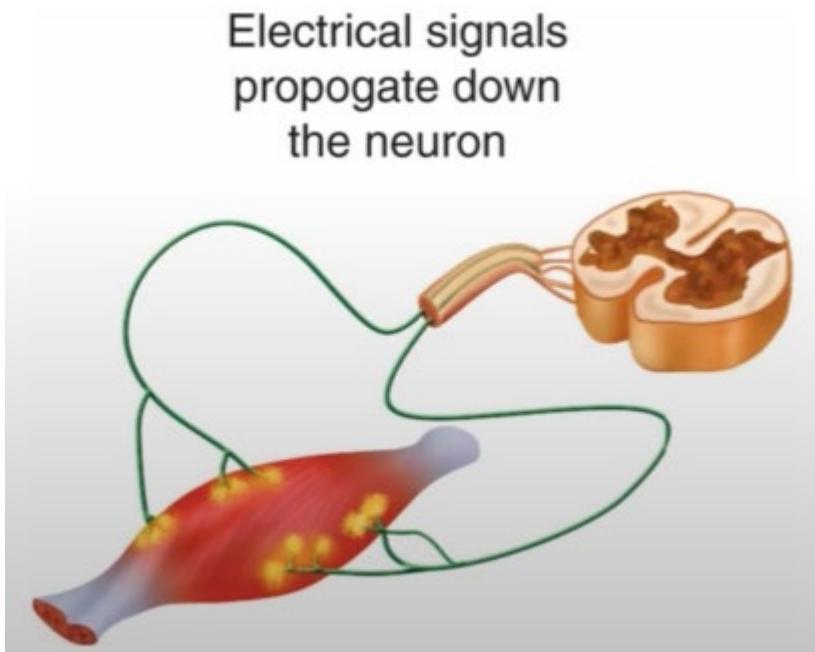
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SDU Robotics

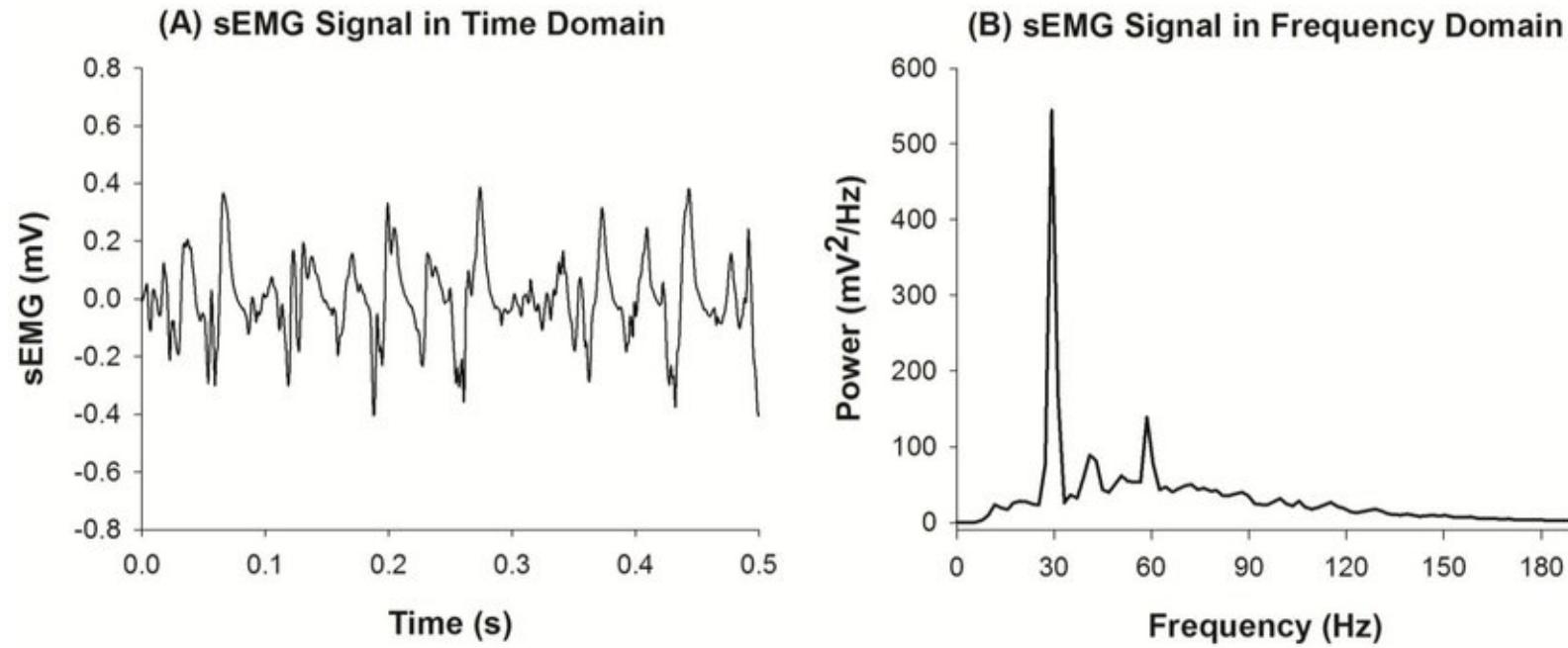
# Topics to be covered in this course

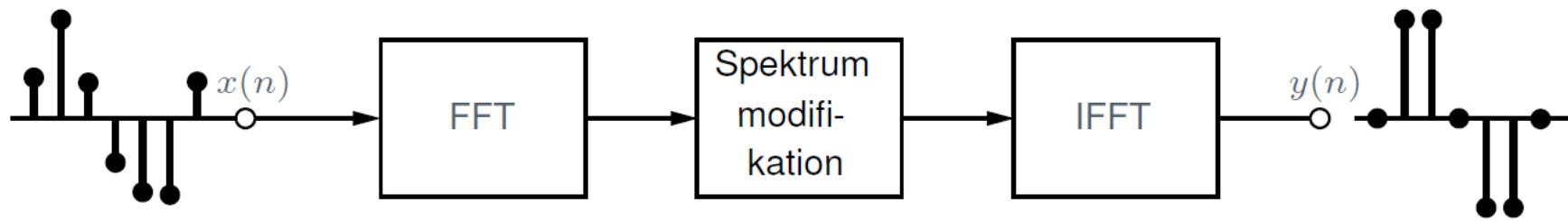
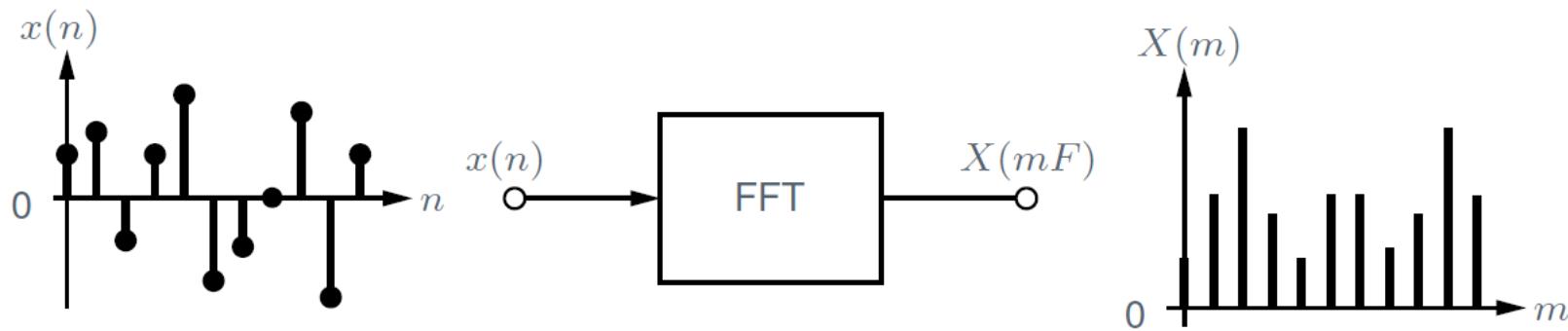
- Sampling and reconstruction
- Aliasing
- Quantization and dynamic range
- Implementation
- Conversion time-frequency domain
- Z transform
- Linear Time Invariant system (LTI)
- System analysis
- Window functions
- Filter design
- Impulse response (FIR and IIR)

# Muscle activity by Electromyography (EMG)

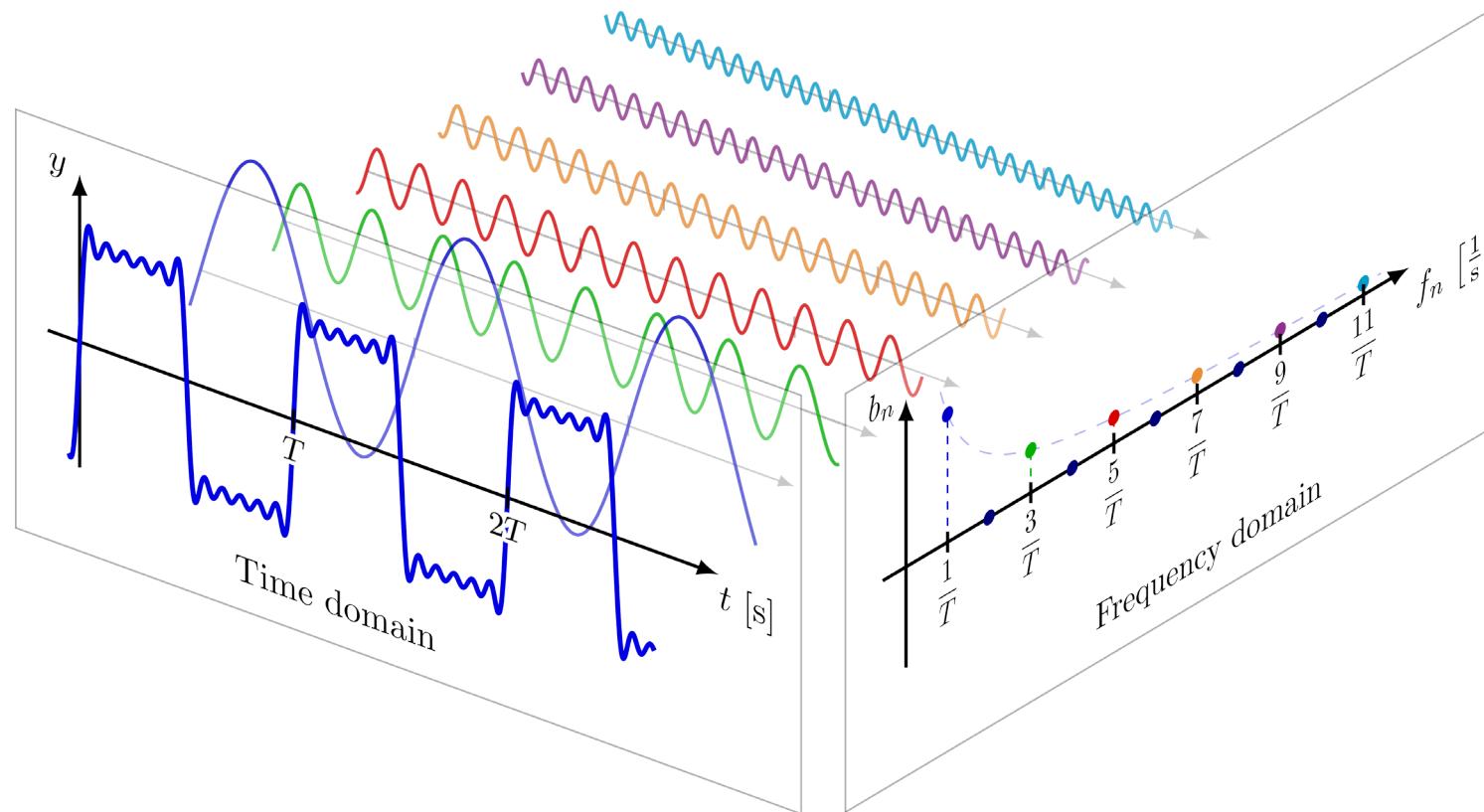


# Convert time domain signal to frequency domain for analysis





# Time – Frequency convert



# Fourier Transform

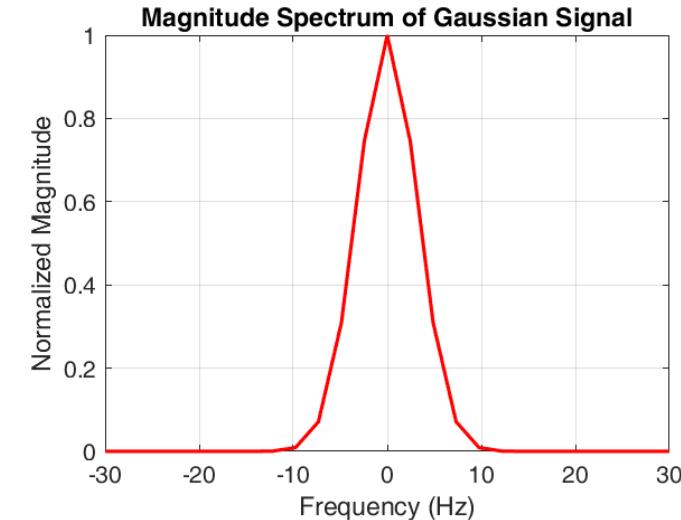
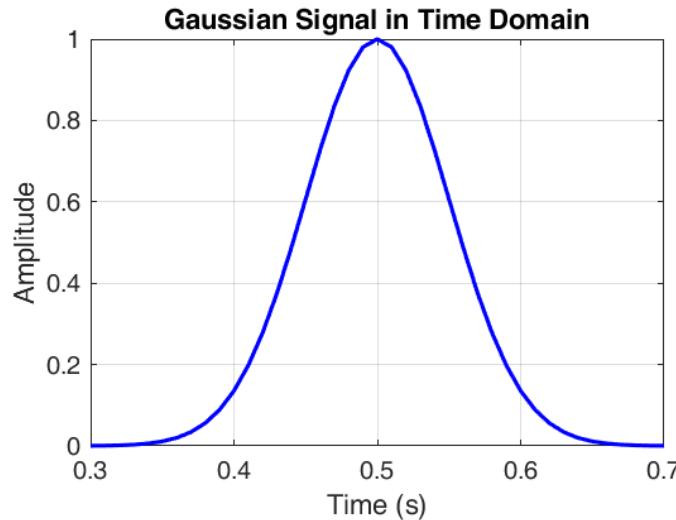
The Fourier transform is a mathematical function that provides frequency spectral analysis.

Given a signal  $x(t)$ , its **Fourier Transform** is defined as

$$X(\omega) = \mathcal{F}(x(t)) = \int_{-\infty}^{\infty} x(t)e^{-j\omega t} dt$$

The signal  $x(t)$  can also be found from the spectrum  $X(\omega)$  via **Inverse Fourier Transform**

$$x(t) = \mathcal{F}^{-1}\{X(\omega)\} = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(\omega)e^{j\omega t} d\omega$$



# Discrete Fourier Transform

$$\omega = 2\pi f$$

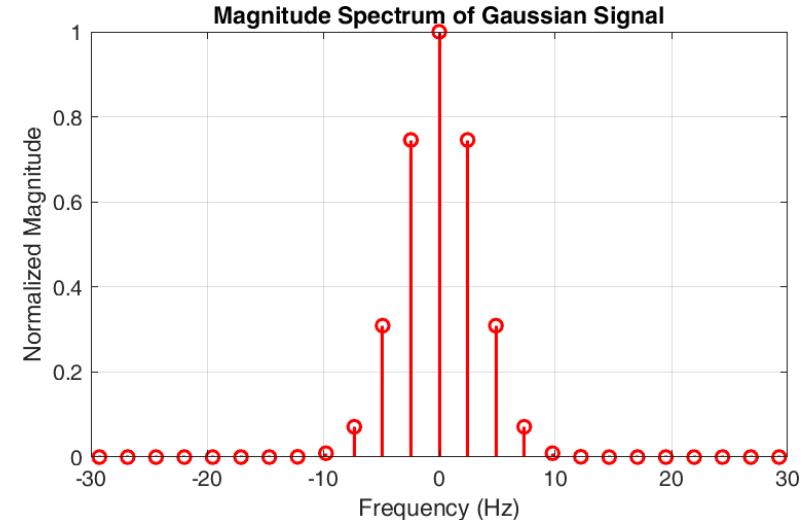
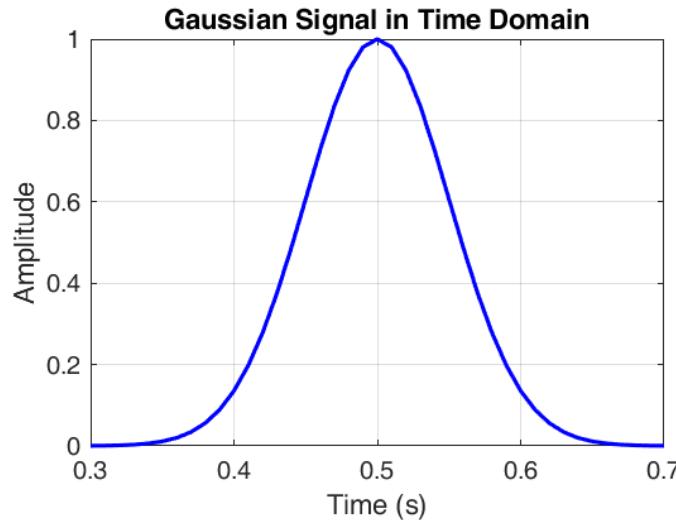
$$X(f) = \mathcal{F}(x(t)) = \int_{-\infty}^{\infty} x(t)e^{-j2\pi ft} dt$$

We must calculate the integral over time t from  $-\infty$  to  $\infty$

For digital signal, we can calculate the **spectrum** function  $X(f)$  by discretizing the frequency range:

$$X(mF) = \int_{-\infty}^{\infty} x(t)e^{-j2\pi mFt} dt$$

$$f = mF$$



$$X(mF) = \int_{-\infty}^{\infty} x(t)e^{-j2\pi mFt} dt$$

Given the sequence  $x(nT)$  which has  $N$  samples,

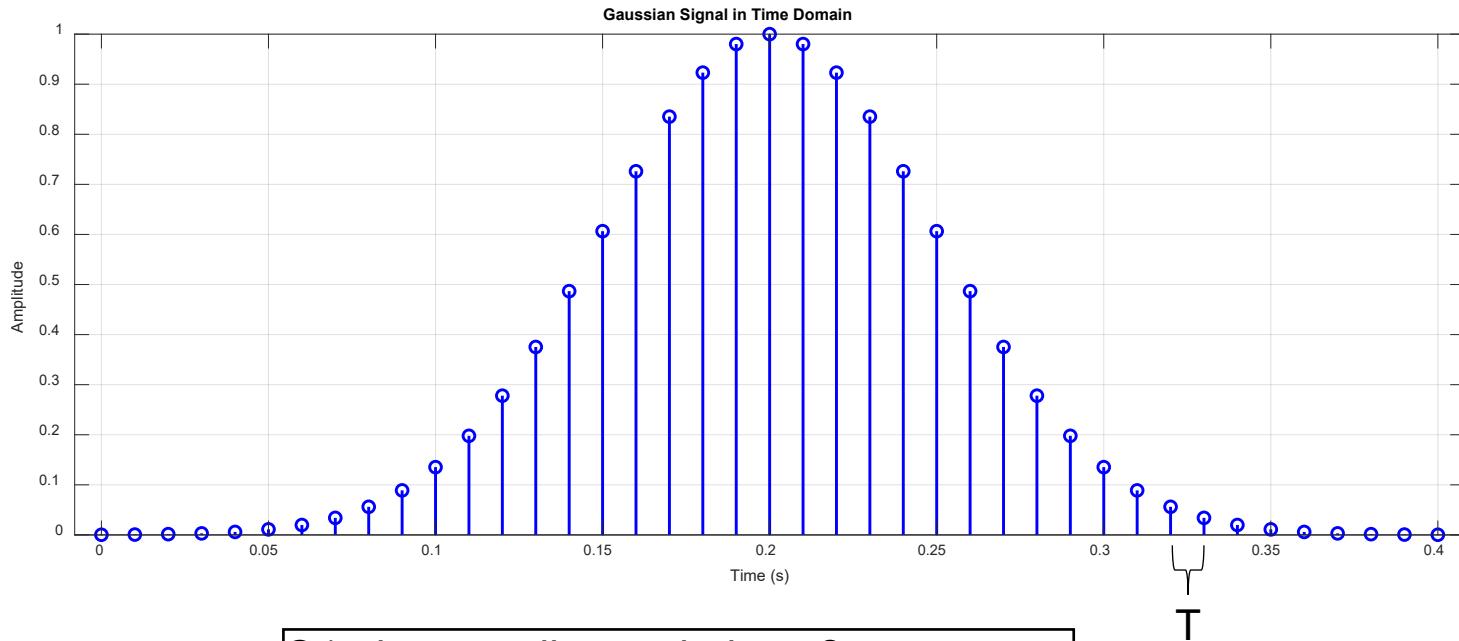
$$X(mF) = T \sum_{n=0}^{N-1} x(nT)e^{-j2\pi mFnT}$$

$$\begin{aligned} \text{Given } f_s &= NF \\ FT &= 1/N \end{aligned}$$

The final DFT can be written as

$$X(m) := \sum_{n=0}^{N-1} x(n)e^{-j\frac{2\pi mn}{N}}$$

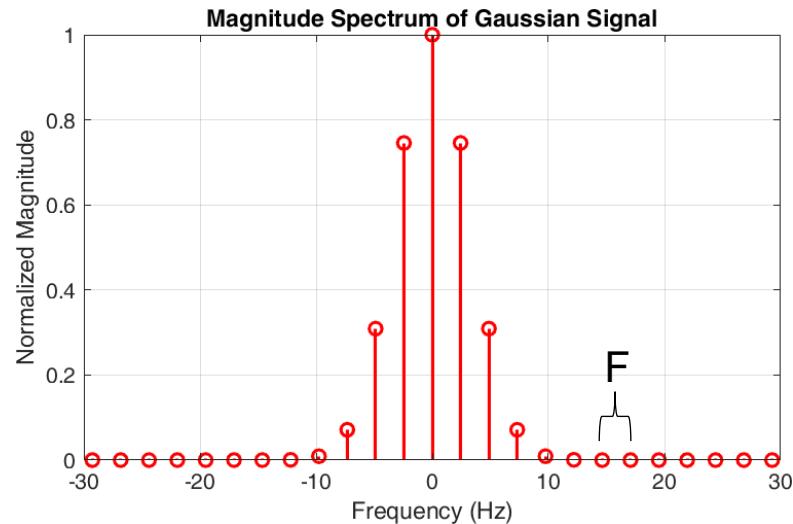
The frequency spacing between the consecutive spectral lines, called the frequency resolution,  $F$ . It also represents the  $m^{\text{th}}$  harmonic.



Q1: the sampling period  $T = ?$

Then the sampling frequency  $f_s = \frac{1}{T} = ?$

In this example,  $N=41$



# Inverse discrete Fourier transformation

Inverse Discrete Fourier Transform (IDFT) is used to calculate the sequence  $x(n)$  giving the spectrum function  $X(f)$ .  
For  $n = 0, 1, \dots, N-1$

$$x(n) = \frac{1}{N} \sum_{m=0}^{N-1} X(m) e^{j2\pi \frac{mn}{N}}$$

It should be noted that both  $X(f)$  and  $x(n)$  are periodic, i.e.

$$x(n) = x(n + kN)$$

$$X(m) = X(m + kN)$$

Because  $x(n)$  is a complex number, it can be rewritten as

$$x(n) = \frac{1}{N} \left( \sum_{n=0}^{N-1} X(m) \cos \left( \frac{2\pi mn}{N} \right) + j \sum_{n=0}^{N-1} X(m) \sin \left( \frac{2\pi mn}{N} \right) \right)$$

We will talk about this again in 'ADC & DAC'!

# Summary

For N-point DFT, the following transformation can be used for calculation:

$x(n)$  is a sequence sampled with interval T

The N-points DFT of  $x(n)$  is given as

$$X(m) = \sum_{n=0}^{N-1} x(n)W_N^{mn}$$

for  $m = 0, 1, \dots, N-1$

Where  $W_N = e^{-j2\pi/N}$

The sequence  $x(n)$  can be found from the spectrum  $X(m)$  as

$$x(n) = \frac{1}{N} \sum_{m=0}^{N-1} X(m)W_N^{-mn}$$

for  $n = 0, 1, \dots, N-1$

# Example - DFT

Considering the following square sequence  $x(n)$



Its spectrum function can be calculated as follows

$$X(m) = \sum_{n=0}^{N-1} x(n) W_N^{mn}$$

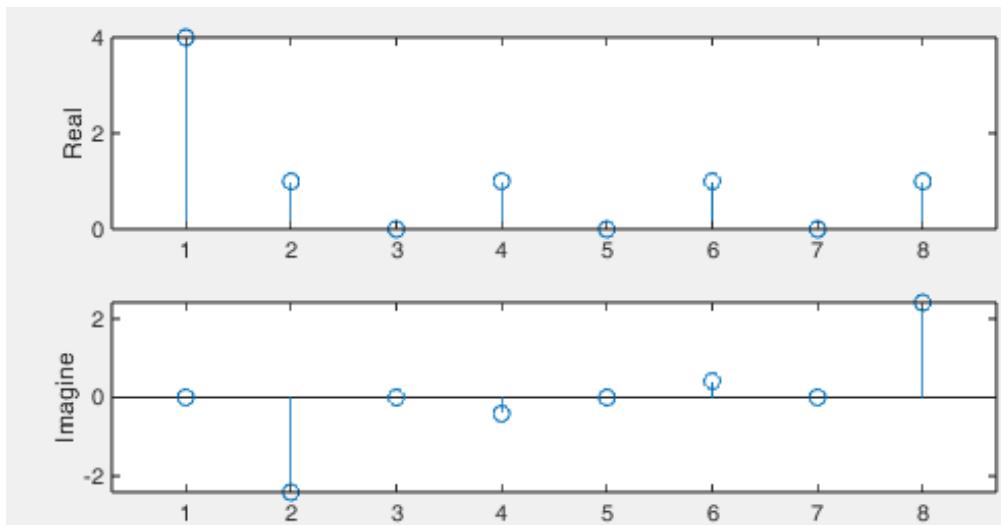
Since N=8, we have

$$X(m) = \sum_{n=0}^7 x(n)W_8^{mn}$$
$$W_8 = e^{-j2\pi/8}$$

Thus, the spectrum function becomes

$$X(m) = \sum_{n=0}^7 x(n) \left( \cos\left(\frac{mn\pi}{4}\right) - j \sin\left(\frac{mn\pi}{4}\right) \right)$$

for  $m = 0, 1, \dots, 7$



%% run Matlab the following

```
x = [1 1 1 1 0 0 0 0]; % input signal  
X = zeros(1,8); % initialize
```

```
for m = 0:7
```

```
for n = 0:7
```

```
X(m+1) = X(m+1) + x(n+1)*(cos(m*n*pi/4) -  
i*sin(m*n*pi/4));
```

```
end
```

```
end
```

```
figure();
```

```
subplot(2,1,1)
```

```
stem(real(X))
```

```
subplot(2,1,2)
```

```
stem(imag(X))
```

% try Matlab inherent function

```
Y= fft(x);
```

% plot Y and check the results

# Example – IDFT

Considering the following sequence in frequency domain

$$X(m) = [10, \quad 2 + j, \quad 0, \quad 2 - j]$$

The iDFT equation is

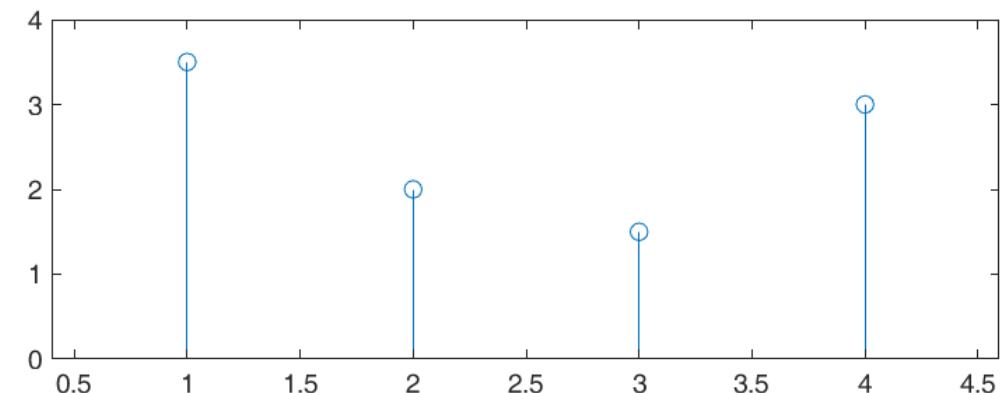
$$x(n) = \frac{1}{N} \sum_{m=0}^{N-1} X(m) e^{j2\pi mn/N}$$

Calculate the Time-domain signal:

$$N = ?$$

```
% again, try Matlab inherent function  
xx = ifft(X);  
% plot xx and check the results
```

```
%%  
X = [10 2+i 0 2-i]; % input signal  
x = zeros(1,4);  
  
for n = 0:3  
for m = 0:3  
x(n+1) = x(n+1) + X(m+1)*exp(i*2*pi*m*n/4);  
end  
end  
x = x / 4;  
stem(x)
```



# Amplitude & phase & power spectrum

The amplitude and phase of  $X(m)$  is

$$A_m = \frac{1}{N} |X(m)| = \frac{1}{N} \sqrt{[Re(X(m))]^2 + [Im(X(m))]^2} \quad \text{and} \quad \angle X(m) = \tan^{-1}\left(\frac{Im(X(m))}{Re(X(m))}\right)$$

Where  $k = 0, 1, 2, \dots, N - 1$

Power spectrum is defined:

$$P(m) = \frac{1}{N^2} |X(m)|^2 = \frac{1}{N^2} ([Re(X(m))]^2 + [Im(X(m))]^2)$$

What we talking about here is in the frequency domain!

# Exercise