

# EE 521: Instrumentation and Measurements

---

Aly El-Osery

Electrical Engineering Department, New Mexico Tech  
Socorro, New Mexico, USA

---

October 26, 2009

- 1 **Bandlimited and Timelimited Signals**
- 2 **Fourier Transform Overview**
- 3 **A Closer Look on DFT**
- 4 **FFT**
  - FFT in MATLAB

## Rect Window

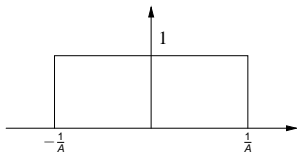
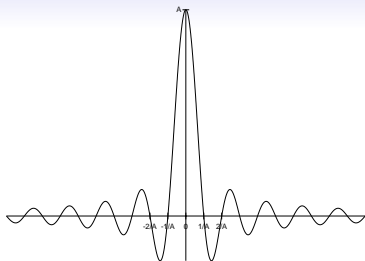


Figure: Fourier transform pair - rectangular window and sinc function

- Fourier transform of a rectangular window is a sinc.
- Inverse Fourier transform of a rectangular window is also a sinc.
- We can only have either timelimited or bandlimited but not both.

## Transform Equations

### DTFT

$$x(n) = \frac{1}{2\pi} \int_{2\pi} X(\omega) e^{j\omega n} d\omega \quad (1)$$

$$X(\omega) = \sum_{n=-\infty}^{\infty} x(n) e^{-j\omega n} \quad (2)$$

## Transform Equations

### DFT

$$X(k) = \sum_{n=0}^{N-1} x(n)e^{-j2\pi kn/N}, \quad k = 0, 1, 2, \dots, N-1 \quad (3)$$

$$x(n) = \frac{1}{N} \sum_{k=0}^{N-1} X(k)e^{j2\pi kn/N}, \quad n = 0, 1, 2, \dots, N-1 \quad (4)$$

## CTFT

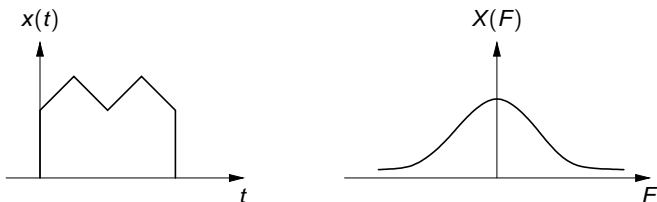


Figure: Continuous Fourier transform of continuous signal

## DTFT

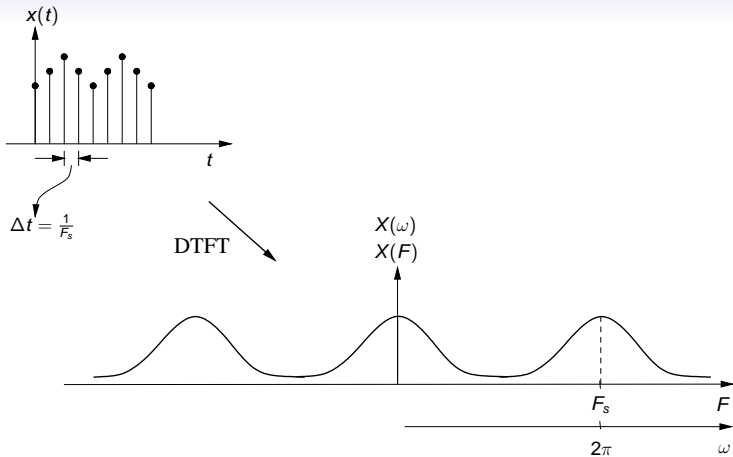


Figure: Discrete time Fourier transform

## DFT

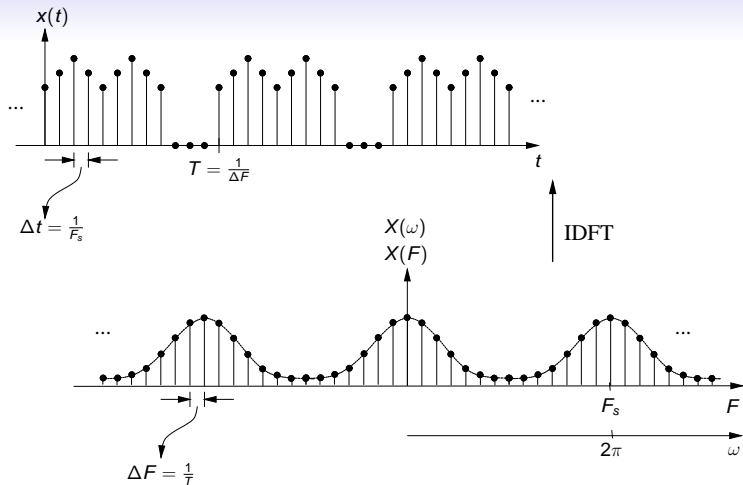


Figure: Discrete Fourier transform



## Limitations

- 1 The number of data points must be finite.
- 2 The computation time required increase as the number of data points increase.
- 3 Frequency resolution is important in determining the signal content.
- 4 Limiting the number of points of a continuous time signal results in *spectral leakage*.

## Number of Points

Assume the number of point in the time domain is  $N_t$  and the number of points in the frequency domain is  $N_F$ .

$$N_t = \frac{T}{\Delta t} = \frac{1/\Delta F}{1/F_s} = \frac{F_s}{\Delta F} = N_F = N \quad (5)$$

## Zero Padding

We can increase the time series sequence by adding zeros and that would not affect it. By doing so the number of points in the time domain,  $N_t$  is increased, and consequently, also is the number of points in the frequency domain,  $N_F$ . Referring to Eq. 5, this means that  $\Delta F$  is decreased.

Zero padding shows more details but not more information

## Window Size

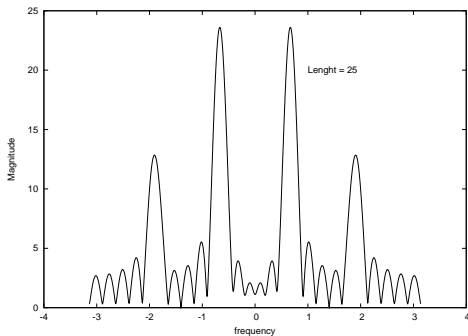


Figure:  $N = 4096$

## Window Size

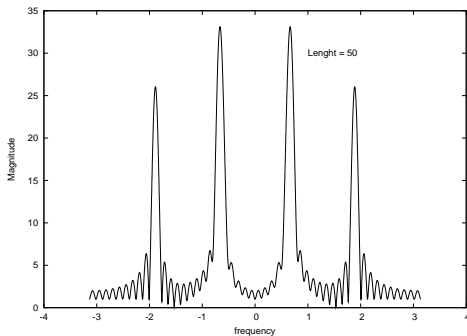


Figure:  $N = 4096$

# Window Size

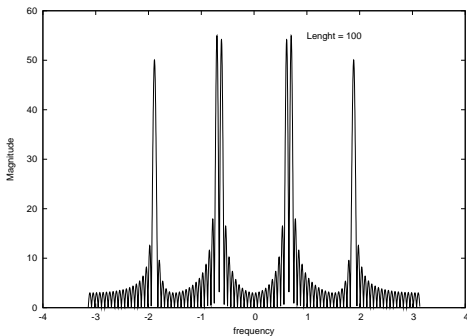


Figure:  $N = 4096$

## Min. Resolvable Resolution

The DTFT of a rectangular window of length  $L$  is given by

$$W(\omega) = \frac{\sin(\omega L/2)}{\sin(\omega/2)} e^{-j\omega(L-1)/2} \quad (6)$$

**To avoid main lobes of overlapping**

$$|\omega_1 - \omega_2| > 2\pi/L \quad (7)$$

## Different Windows

Spectral leakage is due to the sharp cut-off rectangular window. To reduce this effect different windows with smoother roll-off are used. **This is at the cost of wider main lobe which may be undesirable.**



# Hamming Window

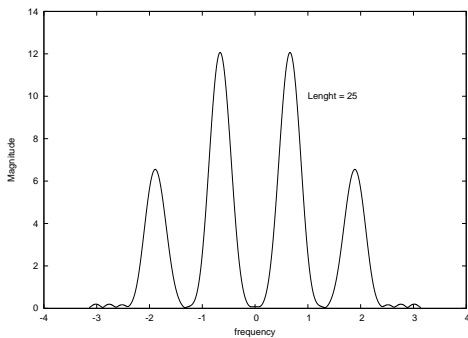


Figure:  $N = 4096$

# Hamming Window

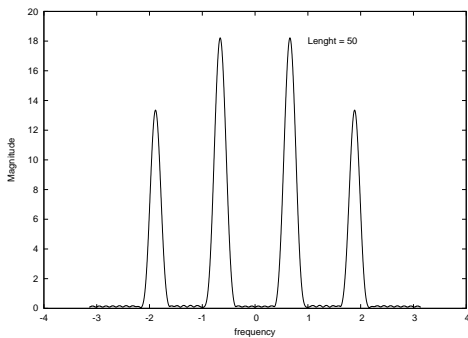


Figure:  $N = 4096$

# Hamming Window

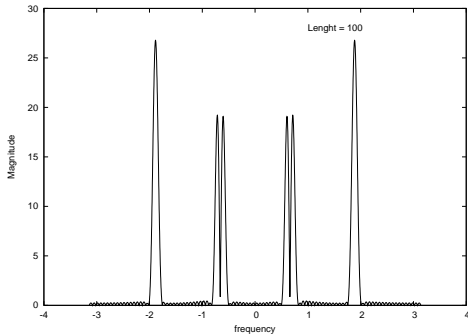


Figure:  $N = 4096$

- An efficient way to compute DFT.
- Direct computation of DFT requires approx.  $N^2$  complex multiplications and  $N^2$  complex additions.
- FFT algorithm requires approximately  $N/2 \log_2 N$ .
- For a 1024 point signal direct computation requires 1,048,576 complex computation versus 5,120 of the FFT.

## Example

Best way to explain that is using an example. Assume that we have a sinusoidal signal that we want to determine its Fourier transform.

```
>> fs=100;  
>> t=0:1/fs:1;  
>> y=cos(2*pi*t)  
>> plot(t,y)  
>> xlabel('t')  
>> ylabel('y(t)')
```

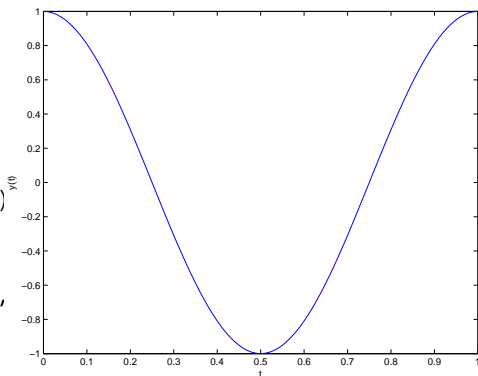


Figure: 1sec sinwave of 1Hz

## Straight FFT

Use the following commands to compute the FFT, find its length and plot the magnitude of  $Y$ .

```
>> Y=fft(y);  
>> length(Y)  
ans =  
    101  
>> plot(abs(Y))
```

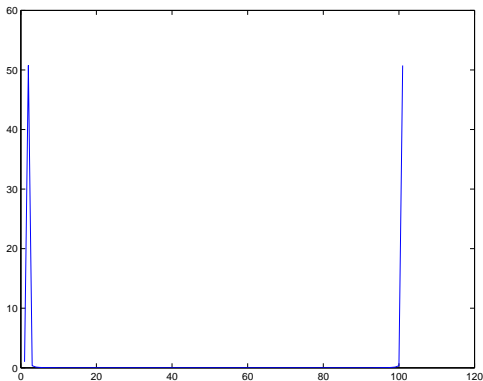


Figure: 101-point FFT

## More Details

We can specify the length of the FFT to be longer by

```
>> N=1024;  
>> Y=fft(y,N);  
>> plot(abs(Y))
```

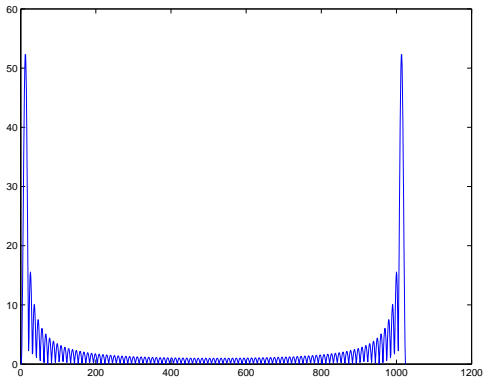


Figure: 1024-point FFT

## fftshift

```
>> plot(fftshift(abs(Y)))
```

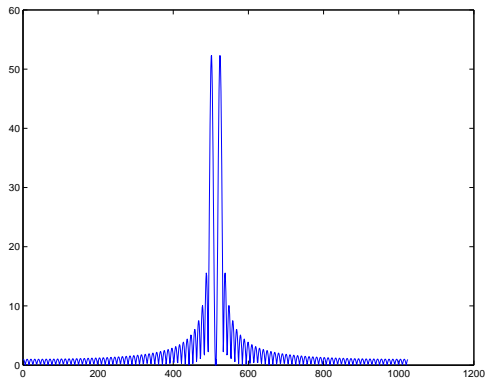


Figure: fftshift



## Axes Mapping

```
>> k=-N/2:N/2-1;  
>> plot(k,fftshift(abs(Y)))
```

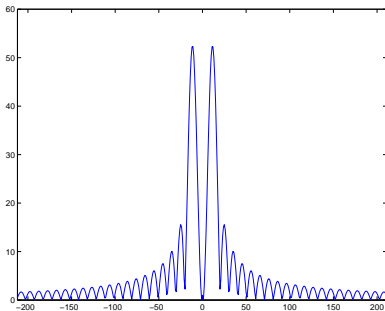


Figure: `fftshift` with  $k = -N/2 : N/2 - 1$

## Axes Mapping

Now we get the axis containing positive and negative values.  
All we have left to do is to map it to actual frequencies.

```
>> plot(k*fs/N,fftshift(abs(Y)))
```

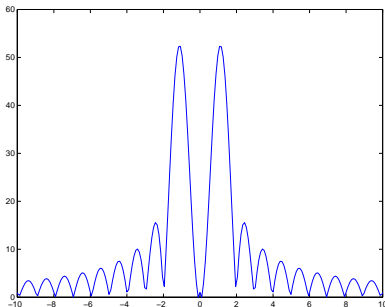


Figure: Frequency mapping

## Window Length

```
>> fs=100;  
>> t=0:1/fs:10;  
>> y=cos(2*pi*t);  
>> N=4096;  
>> Y=fft(y,N);  
>> k=-N/2:N/2-1;  
>> plot(k*fs/N,...  
fftshift(abs(Y))
```

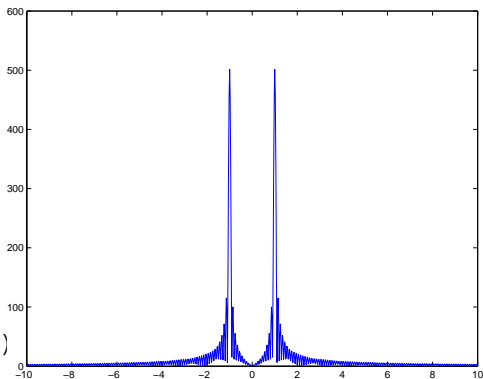


Figure: FFT of a 10sec 1Hz sinwave