

Fig. 1.21 Impulse response of fig. 1.19

Experiment No. 02

Discrete Time Fourier Transform (DTFT)

Basic Theory:

A continuous time signal $f(t)$ called analog signal has a finite amplitude at every instant t . A discrete time signal $x(n)$ called sequence is defined at discrete instants n but $x(n)$ is undefined at any instant between $n+k$ and $n+k+1$. A digital signal is one for which both time and amplitude axis is discrete. Fourier transform is applicable for a discrete time sequence $x(n)$ in a different form called Discrete Fourier Transform (DFT).

For a sequence $x(n)$ in n -domain, its DFT will transform it in m -domain resembles to transformation of time to frequency domain. DFT of $x(n)$ of infinite length is expressed like,

$$X(m\omega) = X(m) = \sum_{n=-\infty}^{\infty} x(n)e^{-jmn\omega} = \sum_{n=-\infty}^{\infty} x(n)e^{-jm(2\pi/N)n} ; \text{ Which resembles to continuous}$$

time Fourier transform of, $X(\omega) = \int_{-\infty}^{\infty} x(t)e^{-j\omega t} dt$.

A comparison between continuous and discrete time Fourier transform is shown in table-2.1.

Table-2.1

Continuous FT	Discrete FT
$X(\omega)$	$X(m)$
$\int_{-\infty}^{\infty}$	$\sum_{n=-\infty}^{\infty}$

t	n
ω	$m\omega$ or m
$x(t)$	$x(n)$
T	N

Now the inverse operation i.e. IDFT can be expressed like,

$$x(n) = \lim_{N \rightarrow \infty} \frac{1}{N} \sum_{m=-\infty}^{\infty} X(m) e^{jm\omega n} = \lim_{N \rightarrow \infty} \frac{1}{N} \sum_{n=-\infty}^{\infty} X(m) e^{jm(2\pi/N)n}$$

For a finite length sequence of length N over $0 \leq n \leq N-1$, DFT and IDFT is expressed like,

$$X(m\omega) = X(m) = \sum_{n=0}^{N-1} x(n) e^{-jm\omega n} = \sum_{n=0}^{N-1} x(n) e^{-jm(2\pi/N)n} \quad \text{and}$$

$$x(n) = \frac{1}{N} \sum_{m=0}^{N-1} X(m) e^{jm\omega n} = \frac{1}{N} \sum_{n=0}^{N-1} X(m) e^{jm(2\pi/N)n}$$

1. Consider a continuous time signal, $x(t) = \sin(2\pi 1000t) + \frac{1}{2} \sin(2\pi 2000t + \frac{3\pi}{4})$.

Determine sampled signal $x(nT_s) = x(n)$, using Matlab, taking $N = 8$ samples at sampling rate, $F_s = 8000\text{Hz}$ (samples/sec).

The sampled signal,

$$x_s(n) = x(nT_s) = \sin(2\pi 1000nT_s) + \frac{1}{2} \sin(2\pi 2000nT_s + \frac{3\pi}{4}); \text{ where } T_s = 1/F_s = 1/8000\text{sec}$$

is the sampling period. Let us use Matlab code to determine sampled sequence $x(n)$.

n=0:1:7;

Ts=1/8000;

xn=sin(2*pi*1000*n*Ts)+0.5*sin(2*pi*2000*n*Ts+3*pi/4);

xn

xn =

0.3536 0.3536 0.6464 1.0607 0.3536 -1.0607 -1.3536 -0.3536

Therefore,

$$x(0) = 0.3536$$

$$x(1) = 0.3536$$

$$x(2) = 0.6464$$

$$x(3) = 1.0607$$

$$x(4) = 0.3536$$

$$x(5) = -1.0607$$

$$x(6) = -1.3536$$

$$x(7) = -0.3536$$

2. Determine DFT of above sequence using Matlab.

$$\mathbf{X} = \text{fft}(\mathbf{xn})$$

$$\mathbf{X} =$$

$$X(0) = 0.0000$$

$$X(1) = -0.0000 - 4.0000i$$

$$X(2) = 1.4142 + 1.4142i$$

$$X(3) = -0.0000 + 0.0000i$$

$$X(4) = -0.0000$$

$$X(5) = -0.0000 - 0.0000i$$

$$X(6) = 1.4142 - 1.4142i$$

$$X(7) = -0.0000 + 4.0000i$$

IDFT of above sequence will retrieve the original sequence $x(n)$ like,

$$\mathbf{Y} = \text{ifft}(\mathbf{X})$$

$$\mathbf{Y} =$$

$$0.3536 \quad 0.3536 \quad 0.6464 \quad 1.0607 \quad 0.3536 \quad -1.0607 \quad -1.3536 \quad -0.3536$$

3. If $x(n) \leftrightarrow X(m)$; both $x(n)$ and $X(m)$ are the vector of length N then their relation can be expressed in a different way like,

$$\mathbf{X} = \mathbf{D}_N \mathbf{x}$$

$$\text{Where } \mathbf{x} = [x(0) \ x(1) \ x(2) \ \dots \ \dots \ \dots \ x(N-1)]^T,$$

$$\mathbf{X} = [X(0) \ X(1) \ X(2) \ \dots \ \dots \ \dots \ X(N-1)]^T,$$

$$\mathbf{D}_N = \begin{bmatrix} 1 & 1 & 1 & \dots & 1 \\ 1 & W_N & W_N^2 & \dots & W_N^{N-1} \\ 1 & W_N^2 & W_N^4 & \dots & W_N^{2(N-1)} \\ \vdots & \vdots & \vdots & \dots & \dots \\ 1 & W_N^{N-1} & W_N^{2(N-1)} & \vdots & W_N^{(N-1)(N-1)} \end{bmatrix} \text{ and } W_N = e^{-j2\pi/N}$$

Determine the \mathbf{D}_N matrix of dimension of 4×4 using Matlab.

$$\mathbf{D} = \text{dftmtx}(4)$$

D =

1.0000	1.0000	1.0000	1.0000
1.0000	0 - 1.0000i	-1.0000	0 + 1.0000i
1.0000	-1.0000	1.0000	-1.0000
1.0000	0 + 1.0000i	-1.0000	0 - 1.0000i

4. Write Matlab code to determine DFT of $x(n) = \{1, 0, 0, 1\}$ hence show the plot of $X(m)$.

```
x=[1 0 0 1];  
y=fft(x);  
subplot(2,1,1)  
stem(abs(y), 'k')  
xlabel('m')  
ylabel('X(m)')  
title('Absolute value of DFT sequence')  
subplot(2,1,2)  
stem(angle(y), 'k')  
xlabel('m')  
ylabel('Angle(X(m))')  
title('Angle of DFT sequence')
```

The result of above code is shown in fig. 2.1.

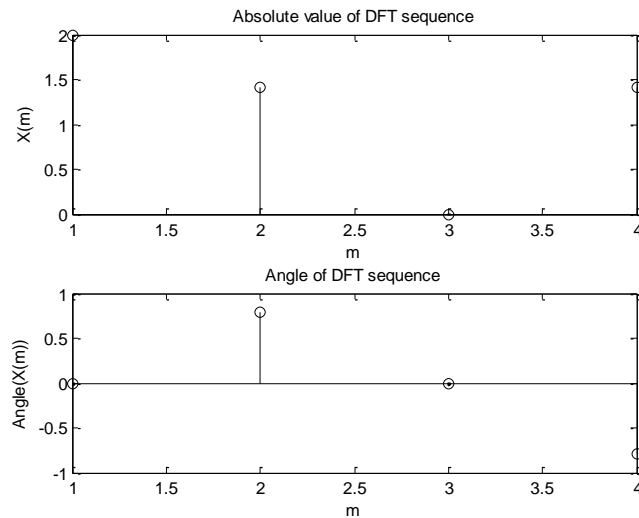
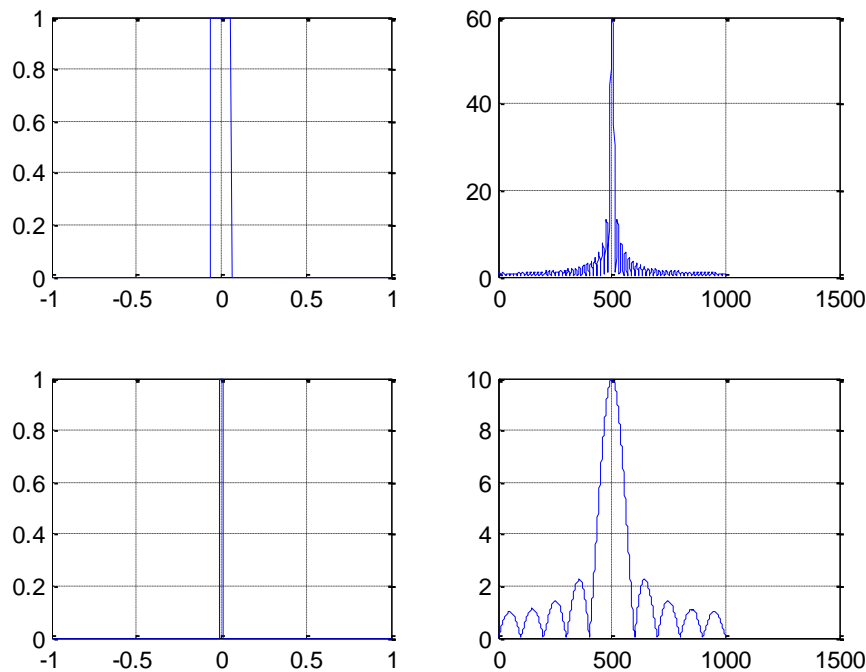


Fig. 2.1 Plot of $|X(m)|$ and $\angle X(m)$

DTFT on a rectangular pulse

```
fs = 500;  
t = -1:1/fs:1;  
x = rectpuls(t,0.12);  
subplot(2,2,1)  
plot(t, x)  
grid on  
y = fft(x);  
y = fftshift(y);  
subplot(2,2,2)  
plot( abs(y))  
grid on  
x = rectpuls(t,0.02);  
subplot(2,2,3)  
plot(t, x)  
grid on  
y = fft(x);  
y = fftshift(y);  
subplot(2,2,4)  
plot( abs(y))  
grid on
```



DTFT on a 3D rectangular pulse

```
x=zeros(32);
```

```
x(12:17)=ones(6,1);
```

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

```
title('2D rectangular pulse')
```

```
mesh(x)
```

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

```
x=fftshift(x);%Dc avlue is at the corner of the array
```

```
%let us move it at the middle
```

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

```
mesh(abs(x))
```

```
title('2D sinc in frequency domain')
```

```

x=zeros(32);
x(12:17,12:17)=ones(6);
subplot(2,2,3)
title('3D rectangular pulse')
mesh(x)
x=fft2(x);
x=fftshift(x);%Dc avlue is at the corner of the array
%let us move it at the middle
subplot(2,2,4)
mesh(abs(x))
title('3D sinc in frequency domain')

```

