



**VINAYAKA MISSION'S  
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**VINAYAKA MISSION'S  
KIRUPANANDA VARIYAR  
ENGINEERING COLLEGE**

**VINAYAKA MISSION'S KIRUPANANDA VARIYAR ENGINEERING COLLEGE, SALEM**  
(A Constituent College of Vinayaka Mission's Research Foundation, Deemed to be University, Salem)  
(An ISO 9001:2000 Certified, NAAC Accredited and AICTE Approved)  
NH-47, Sankari Main Road, Periya Seeragapadi, Salem – 636 308

<b>DEPARTMENT OF ELECTRONICS AND COMMUNICATION ENGINEERING</b>
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\_\_\_\_\_ **Laboratory Record**

University Reg. No. \_\_\_\_\_

Name: \_\_\_\_\_

Batch: \_\_\_\_\_

Course Name : \_\_\_\_\_ Course Code : \_\_\_\_\_

**CERTIFIED THAT THIS BONAFIDE RECORD OF WORK DONE BY**

**Mr / Ms** \_\_\_\_\_

**Staff-in-charge**

**Head Of the Department**

Salem: 636308

Date: \_\_\_\_\_

**Submitted to the  
Vinayaka Missions Research Foundation (Deemed to be University)  
Practical Examination held on \_\_\_\_\_ 20**

**Internal Examiner**

**External Examiner**

[illegible]

**Exp No: 1**

**Date:**

## **GENERATION OF SIGNALS**

**Aim:**

To write MATLAB programs to generate the following signals

1. Unit impulse function
2. Unit step function.
3. Unit ramp
4. Exponential function

**Essentials required:**

Hardware: IBM PC or compatible

Software: MATLAB v5.1 or higher

**Program:**

```
close all;

clear all;

n=0:.1:5;

% RAMP FUNCTION

figure(1);

subplot(3,3,7);

plot(n);

title('RAMP FUNCTION');

xlabel('time(seconds)');

ylabel('Amplitude');

% IMPLUSE FUNCTION

n=0:.2:5;

figure(1);

subplot(3,3,8);

stem(0,1);

title('IMPLUSE FUNCTION');
```

```

xlabel('time(seconds)');
ylabel('Amplitude');
% UNIT FUNCTION
N=21;
X=ones(1,N);
n=0:1:N-1;
figure(1);
subplot(3,3,9);
stem(n,X);
title('UNIT FUNCTION');
xlabel('time(seconds)');
ylabel('Amplitude');
% EXPONENTIAL FUNCTION
X2=exp(n);
figure(2);
subplot(2,2,1);
plot(n,X2);
title('EXPONENTIAL FUNCTION');
xlabel('time(seconds)');
ylabel('Amplitude');
X3=exp(-n);
figure(2);
subplot(2,2,2);
plot(n,X3);
title('EXPONENTIAL FUNCTION');
xlabel('time(seconds)');
ylabel('Amplitude');
X4=-exp(n);
figure(2);
subplot(2,2,3);

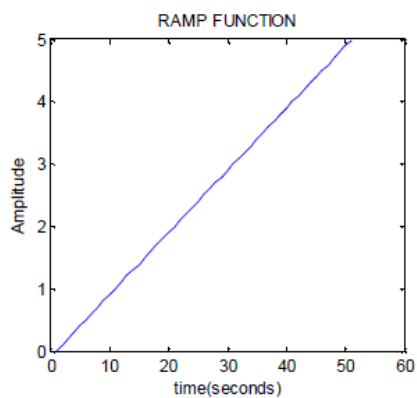
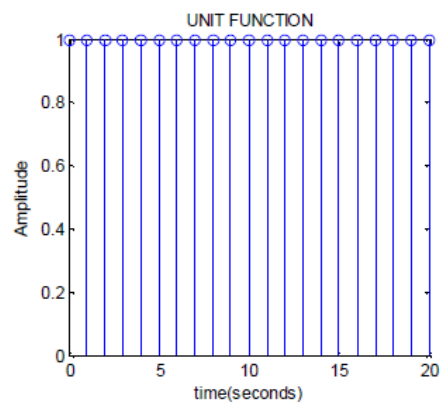
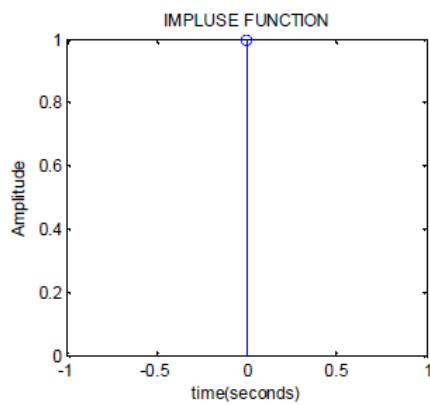
```

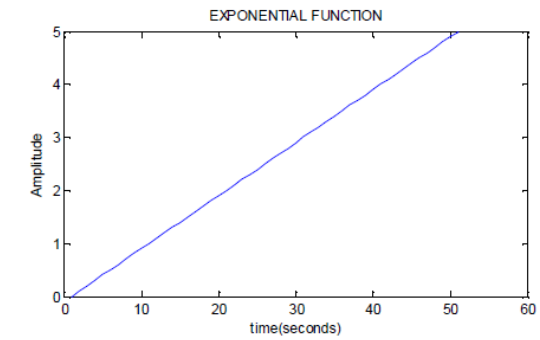
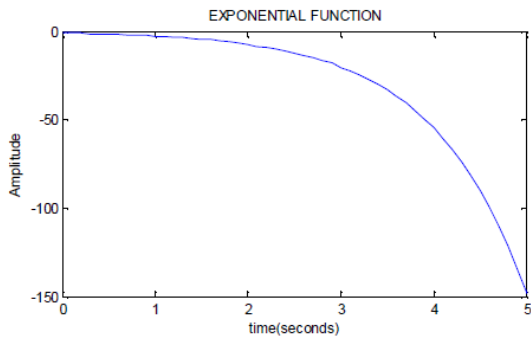
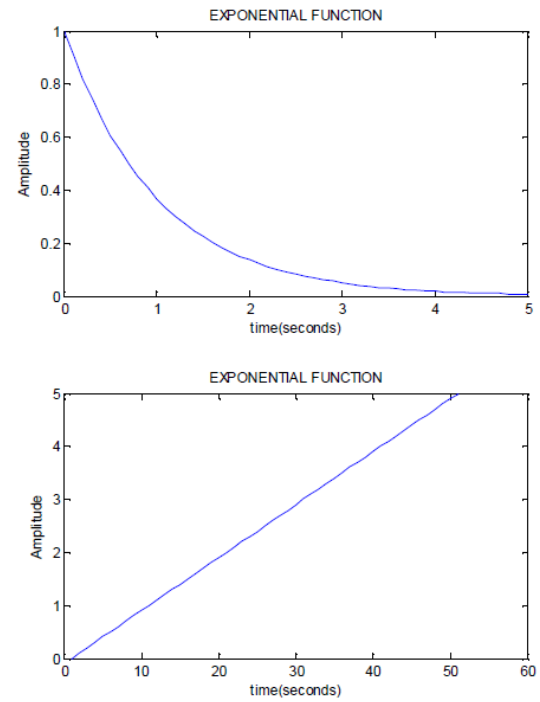
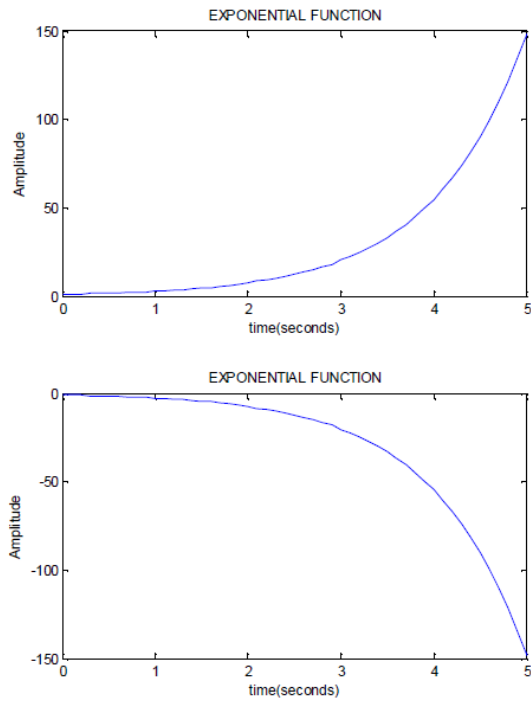
```

plot(n,X4);
title('EXPONENTIAL FUNCTION');
xlabel('time(seconds)');
ylabel('Amplitude');
X5=-exp(-n);
figure(2);
subplot(2,2,4);
plot(n);
title('EXPONENTIAL FUNCTION');
xlabel('time(seconds)');
ylabel('Amplitude');

```

### Output:





## Result:

The MATLAB programs to generate various sequences and waves are written and the results are plotted.

**Exp No: 2**

**Date:**

## **LINEAR CONVOLUTION**

**Aim:**

To write MATLAB programs for the following:

1. Linear convolution

**Essentials required:**

Hardware: IBM PC or compatible

Software : MATLAB v5.1 or higher

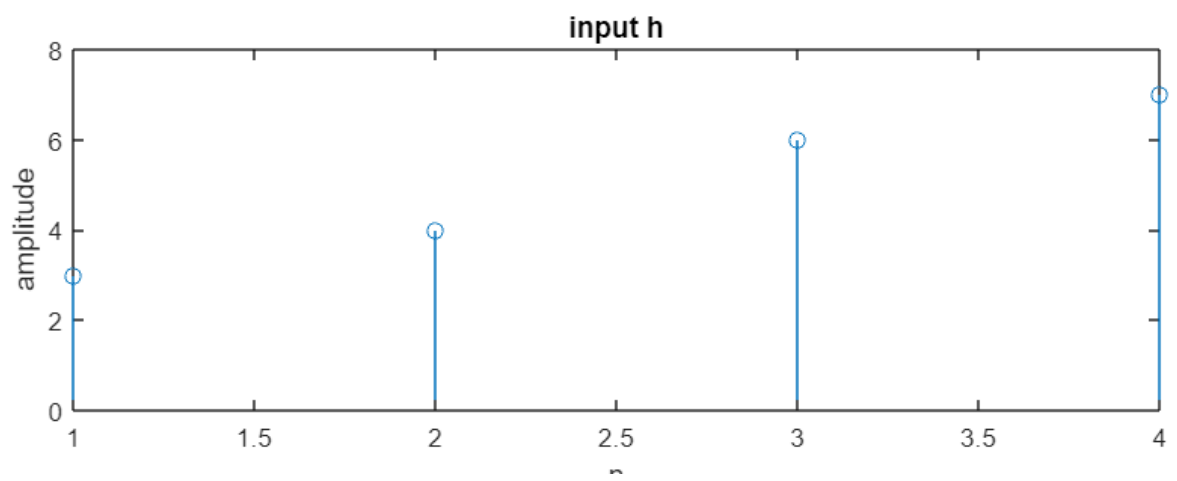
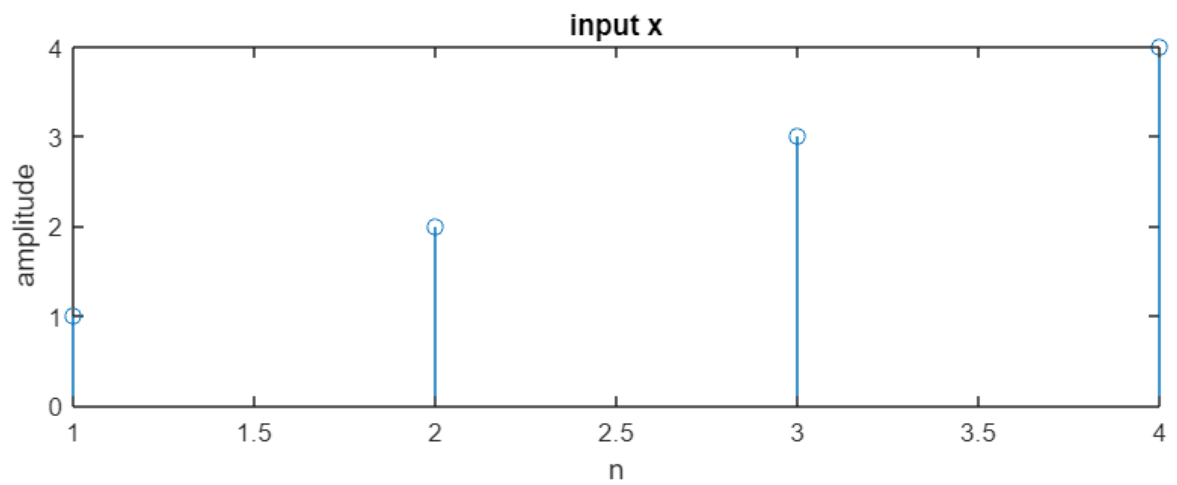
**Program:**

```
clc;
close all;
clear all;
x=input('enter the 1st sequence');
h=input('enter the 2nd sequence');
y=conv(x,h);
subplot(2,1,1);
stem(x);
title('input x');
ylabel('amplitude');
xlabel('n');
subplot(2,1,2);
stem(h);
title('input h');
ylabel('amplitude');
xlabel('n');
figure(2);
stem(y);
title('output y');
```

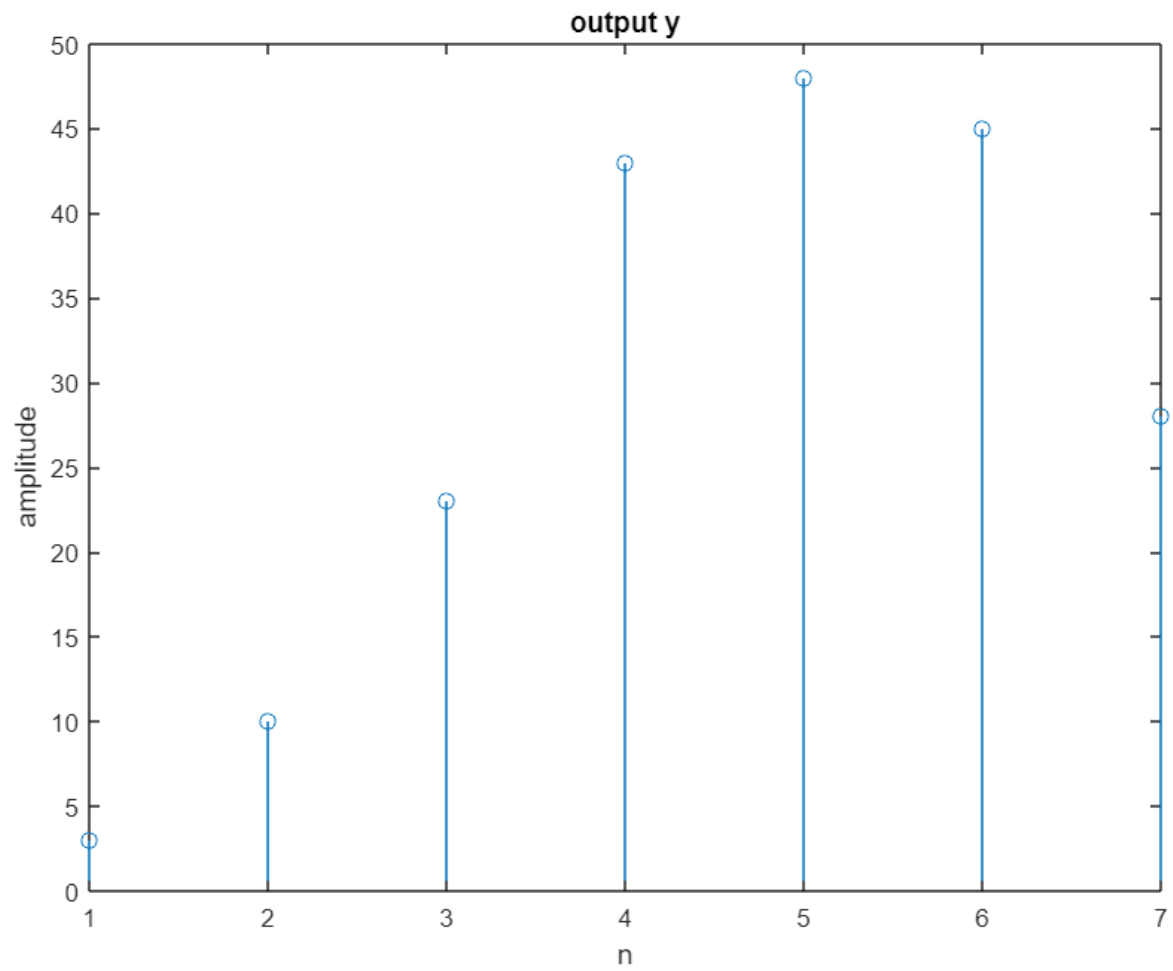
```
ylabel('amplitude');
```

```
xlabel('n');
```

```
disp('the resultant signal');
```







**Result:**

The MATLAB program to implement linear convolution is written and the results are plotted.

**Expt No: 3**

**Date:**

## **CIRCULAR CONVOLUTION**

**Aim:**

To write MATLAB programs for the Circular convolution.

**Essentials required:**

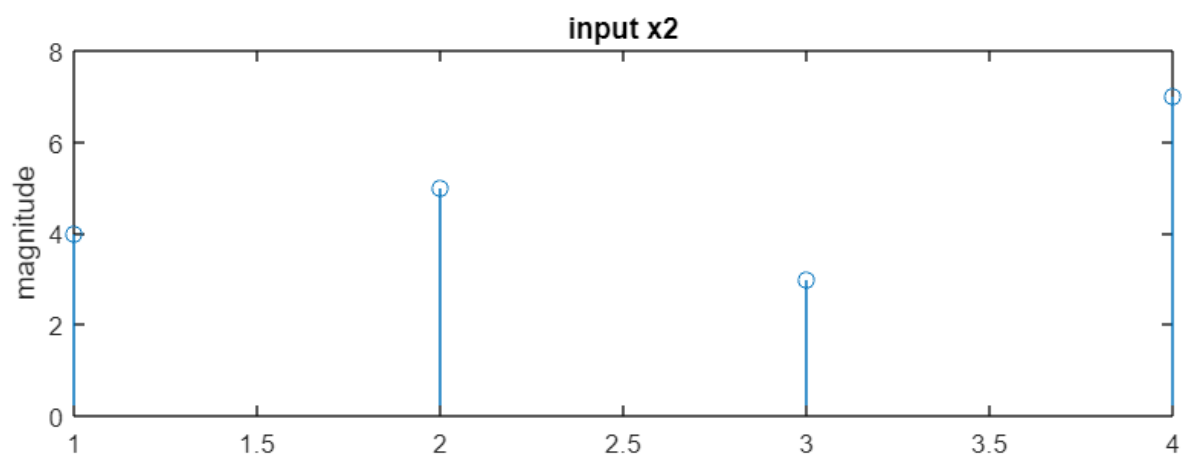
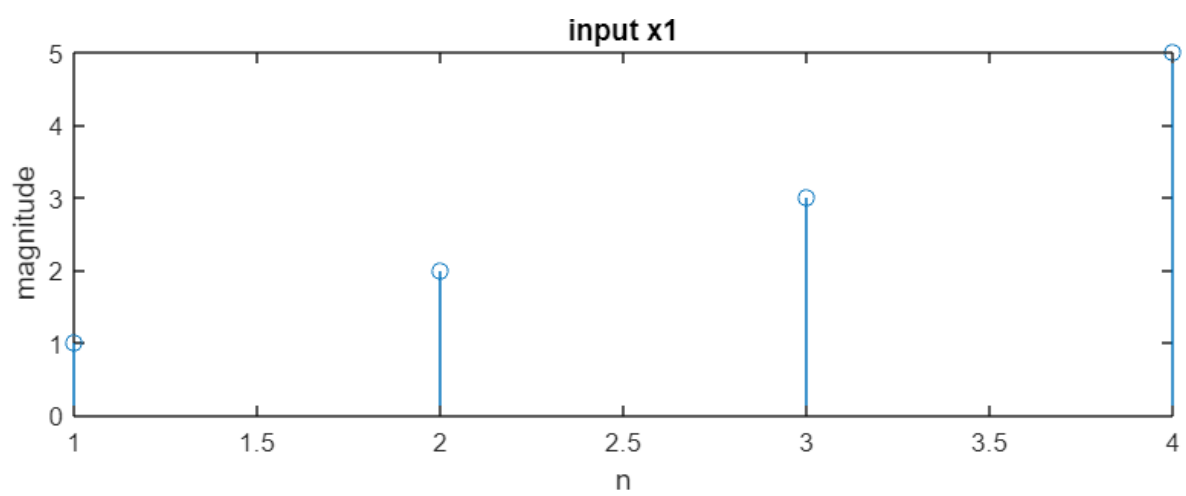
Hardware: IBM PC or compatible

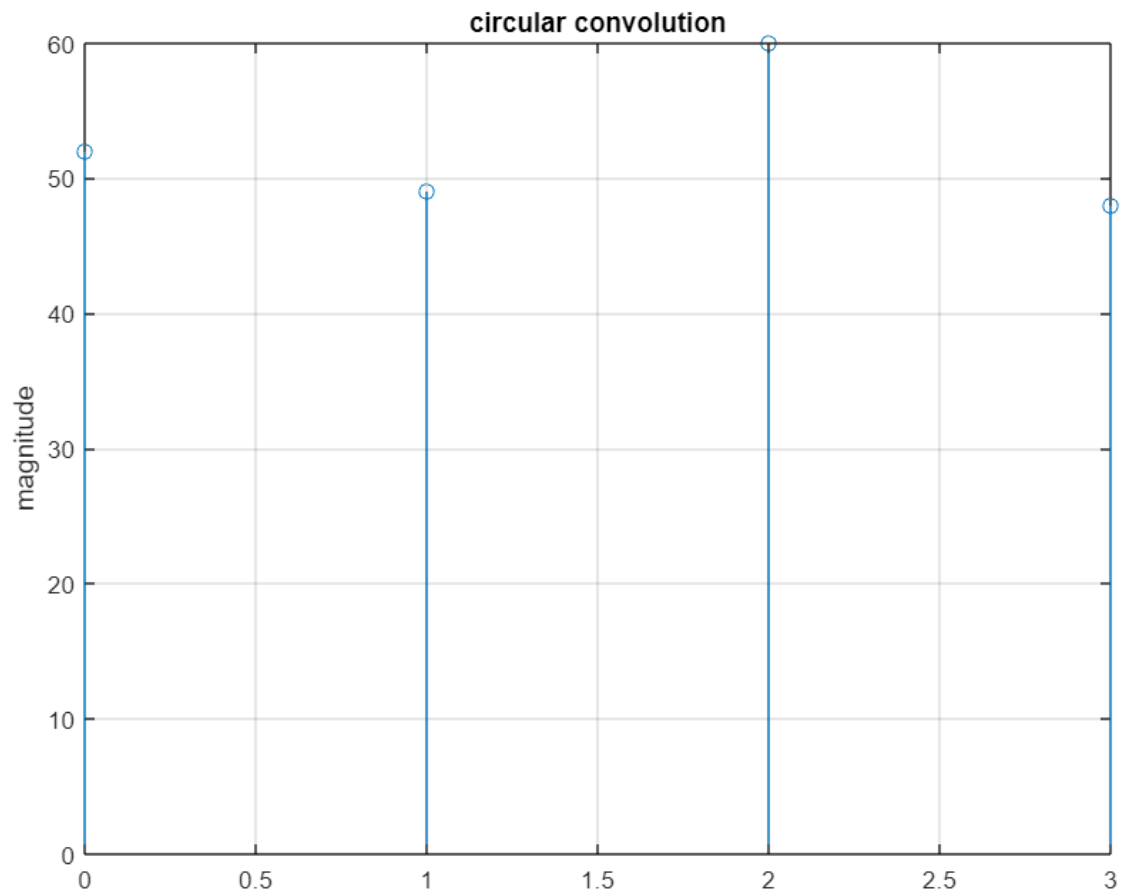
Software: MATLAB v5.1 or higher

**Program:**

```
clc;
x1=input('enter the first sequence');
x2=input('enter the second sequence');
subplot(2,1,1);
stem(x1);
title('input x1');
ylabel('magnitude');
xlabel('n');
subplot(2,1,2);
stem(x2);
title('input x2');
ylabel('magnitude');
xlabel('n');
n=max(length(x1),length(x2));
x1=fft(x1,n);
x2=fft(x2,n);
y=x1.*x2;
yc=ifft(y,n);
disp('circular convolution:');
```

```
disp(yc);  
n=0:1:n-1;  
figure(2);  
stem(n,yc);grid;  
xlabel('n');  
ylabel('magnitude');  
title('circular convolution');
```





**Result:**

The MATLAB programs to implement circular convolution is written and the results are plotted.

**Expt No:4**

**Date:**

## **ANALOG CHEBYSHEV FILTERS AND APPLY BILINEAR TRANSFORMATION**

**Aim:**

To design analog Chebyshev filters and apply bilinear transformation

**Essentials required:**

Hardware: IBM PC or compatible

Software: MATLAB v5.1 or higher

**Program:**

```
clc; % clear screen
clear all; % clear work space
close all; % close all figure windows
fp = input('Enter the Pass band frequency in Hz = '); % input specifications
fs = input('Enter the Stop band frequency in Hz = ');
Fs = input('Enter the Sampling frequency in Hz = ');
Ap = input(' Enter the Pass band ripple in db:');
As = input('Enter the Stop band ripple in db:');
wp=2*pi*fp/Fs; % Analog frequency
ws=2*pi*fs/Fs;
Up = 2*tan(wp/2); % Prewrapped frequency
Us = 2*tan(ws/2);
[N,wn]= cheb1ord (Up,Us,Ap,As,'s'); %Calculate order and cutoff freq
disp('order of the filter N = ');
disp(N);
disp('Normalized cut off frequency = ');
disp(wn);
[num, den] = cheby1(N, Ap,wn,'s');
[b,a] = bilinear(num,den,1);
freqz(b,a,512,Fs);
printsys(b,a,'z');
```

### OUTPUT:

enter the Pass band edge frequency in Hz = 100

enter the stop band frequency in Hz = 500

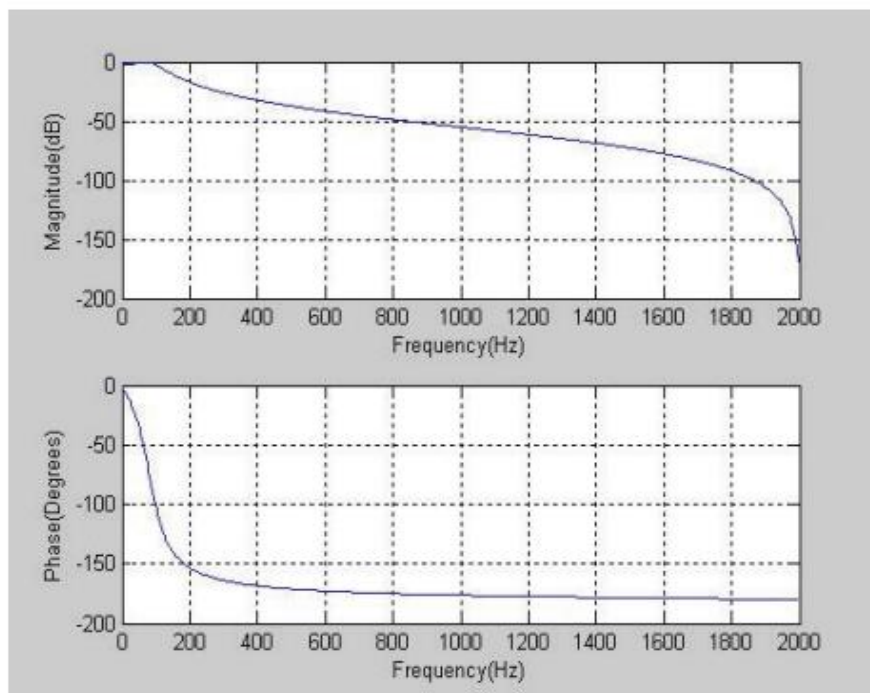
enter the sampling frequency in Hz = 4000

enter the pass band ripple n db = 2

enter the stop band attenuation in db =

20 order of the filter N = 2

Normalised cutoff frequency = 0.1574



### Result:

The MATLAB programs to implement analog Chebyshev filters and apply bilinear transformation is written and the results are plotted.

**Exp No: 5**

**Date:**

## **BUTTERWORTH FILTERS AND APPLY BILINEAR TRANSFORMATION**

**Aim:**

To design analog Butterworth filters and apply bilinear transformation

**Essentials required:**

Hardware: IBM PC or compatible

Software: MATLAB v5.1 or higher

**Program:**

```
clc; % clear screen
clear all; % clear screen
close all; % close all figure windows
fp = input('Enter the Pass band frequency in Hz = '); % input specifications
fs = input('Enter the Stop band frequency in Hz = ');
Fs = input('Enter the Sampling frequency in Hz = ');
Ap = input(' Enter the Pass band ripple in db:');
As = input('Enter theStop band ripple in db:');
wp=2*pi*fp/Fs; % Analog frequency
ws=2*pi*fs/Fs;
Up = 2*tan(wp/2);% Prewrapped frequency
Us = 2*tan(ws/2);
[n,wn]= buttord (Up,Us,Ap,As,'s'); %Calculate order and cutoff freq
disp('Order of the filter N =');
disp(n);
disp('Normalized cut off frequency = ');
disp(wn);
[num, den] = butter(n,wn,'s'); % analog filter transfer
[b,a] = bilinear(num, den,1); % conversion of analog filter to digital filter
freqz(b,a,512,Fs); % frequency response of the filter
printsys(b,a,'z'); % print the H(z) equation obtained on screen
```

**OUTPUT:**

enter the Pass band edge frequency in Hz = 500

enter the stop band frequency in Hz = 750

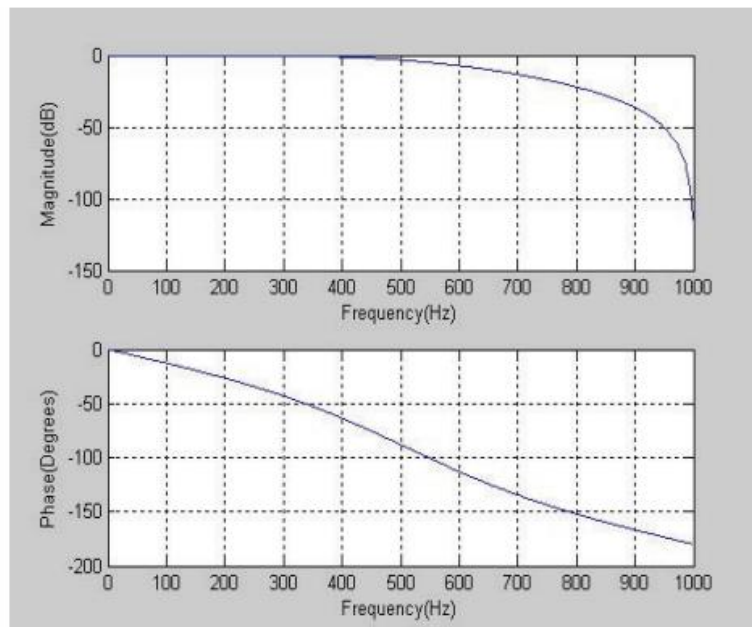
enter the sampling frequency in Hz = 2000

enter the pass band ripple n db = 3.01

enter the stop band attenuation in db =

15 order of the filter N = 2

Normalised cutoff frequency = 2.052

**Result:**

The MATLAB program to implement analog Butterworth filters and apply bilinear transformation is written and the results are plotted.



**Exp No: 6**

**Date:**

## **ANALOG CHEBYSHEV FILTERS AND APPLY IMPULSE INVARIANCE TRANSFORMATION**

**Aim:**

To design analog Chebyshev filters and apply impulse invariance transformation.

Consider Problem: Design a Chebyshev digital IIR using impulse invariant transformation by taking  $T=1$  sec to satisfy the following specifications:

$$0.9 \leq |H(e^{j\omega})| \leq 1.0; \text{ for } 0 \leq \omega \leq 0.28\pi$$
$$|H(e^{j\omega})| \leq 0.24; \text{ for } 0.5\pi \leq \omega \leq \pi$$

**Essentials required:**

Hardware: IBM PC or compatible

Software: MATLAB v5.1 or higher

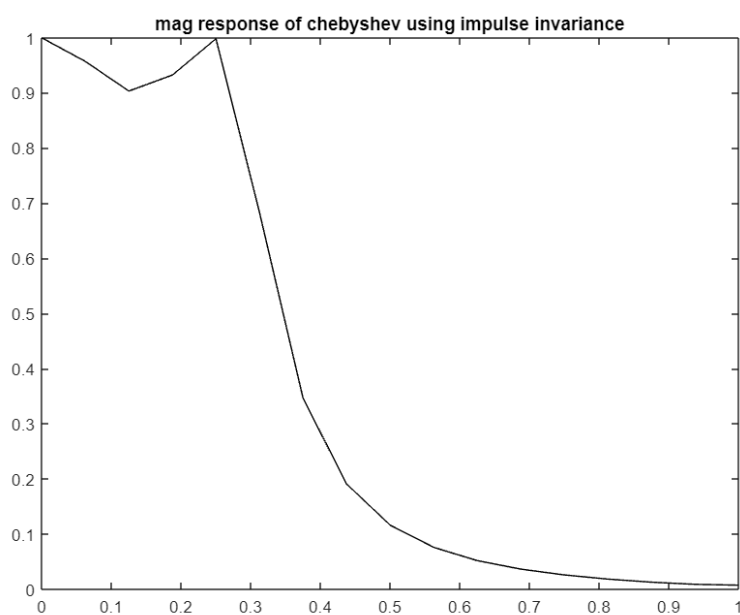
**Program:**

```
close all;
clear all;
clc
ap=0.9;
as=0.24;
p_d=0.28*pi;
s_d=0.5*pi;
t=1;
pass_attenuation=-20*log10(ap);
stop_attenuation=-20*log10(as);
p_a=p_d/t;
s_a=s_d/t;
[n,cf]=cheb1ord(p_a,s_a,pass_attenuation,stop_attenuation,'s');
[bn,an]=cheby1(n,pass_attenuation,1, 's');
hsn=tf(bn,an);
[b,a]=cheby1(n,pass_attenuation,cf,'s');
```

```

hs=tf(b,a);
[num,den]=impinvar(b,a,1/t);
hz=tf(num,den,t);
w=0:pi/16:pi;
hw=freqz(num,den,w);
hw_mag=abs(hw);
plot(w/pi,hw_mag,'k');
title('mag response of chebyshev using impulse invariance');

```



### Result:

The MATLAB program to design analog Chebyshev filters and apply impulse invariance transformation is written and the results are plotted.

**Exp No: 7**

**Date:**

## **ANALOG BUTTERWORTH FILTERS AND APPLY IMPULSE INVARIANCE TRANSFORMATION**

**Aim:**

To design analog butterworth filters and apply impulse invariance transformation

Consider a problem: Design a Butterworth IIR Filter using impulse invariance Transformation by taking  $T=1$  sec and

$$0.707 \leq |H(e^{j\omega})| \leq 1.0; \text{ for } 0 \leq \omega \leq 0.3\pi$$
$$|H(e^{j\omega})| \leq 0.2; \text{ for } 0.75\pi \leq \omega \leq \pi$$

**Essentials required:**

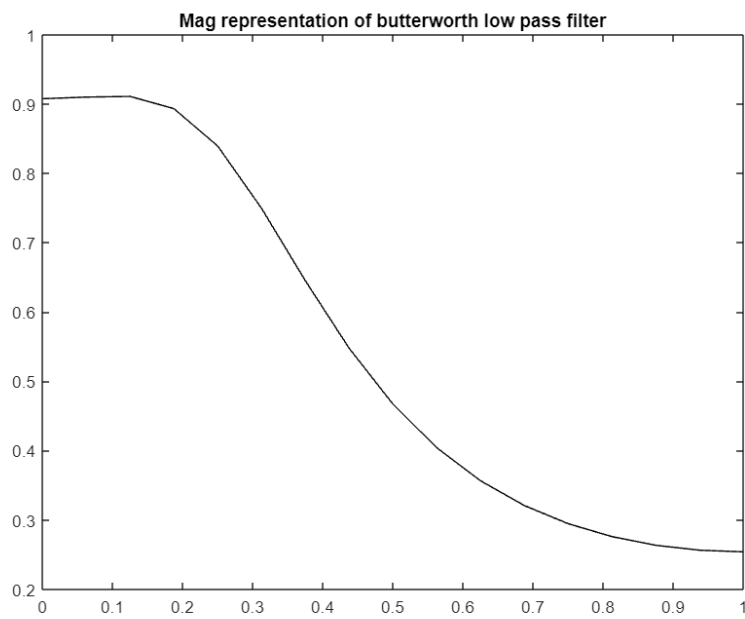
Hardware: IBM PC or compatible

Software: MATLAB v5.1 or higher

**Program:**

```
clear all;
clc;
Ap=0.707;
As=0.2;
omega_p=0.3*pi;
omega_s=0.75*pi;
T=1;
Pass_attenuation=-20*log10(Ap);
Stop_attenuation=-20*log10(As);
P_a=omega_p/T;
S_a=omega_s/T;
[N,CF]=buttord(P_a,S_a,Pass_attenuation,Stop_attenuation, 's');
[Bn,An]=butter(N,1,'s');
Hsn=tf(Bn,An);
[B,A]=butter(N,CF,'s');
HS=tf(B,A);
[num,den]=impinvar(B,A,1/T);
```

```
Hz=tf(num,den,T);  
w=0:pi/16:pi;  
Hw=freqz(num,den,w);  
Hw_mag=abs(Hw);  
plot(w/pi,Hw_mag,'k');  
title('Mag representation of butterworth low pass filter');
```



### Result:

The MATLAB program to design analog butterworth filters and apply impulse invariance transformation written and the results are plotted.

**Exp No: 8**

**Date:**

## **FIR FILTERS USING FOURIER SERIES METHOD AND FREQUENCY SAMPLING METHODS**

**Aim:**

To design a FIR filter using Fourier series method and frequency sampling methods

**Essentials required:**

Hardware: IBM PC or compatible

Software: MATLAB v5.1 or higher

**Program:**

```
% fir filter
```

```
clc;
```

```
clear all;
```

```
close all;
```

```
% low pass filter
```

```
n=50;
```

```
wn=0.5;
```

```
b=fir1(n,wn);
```

```
fvtool(b,1);
```

```
% high pass filter
```

```
n=50;
```

```
wn=0.5;
```

```
b=fir1(n,wn,'high');
```

```
fvtool(b,1);
```

```
% band pass filter
```

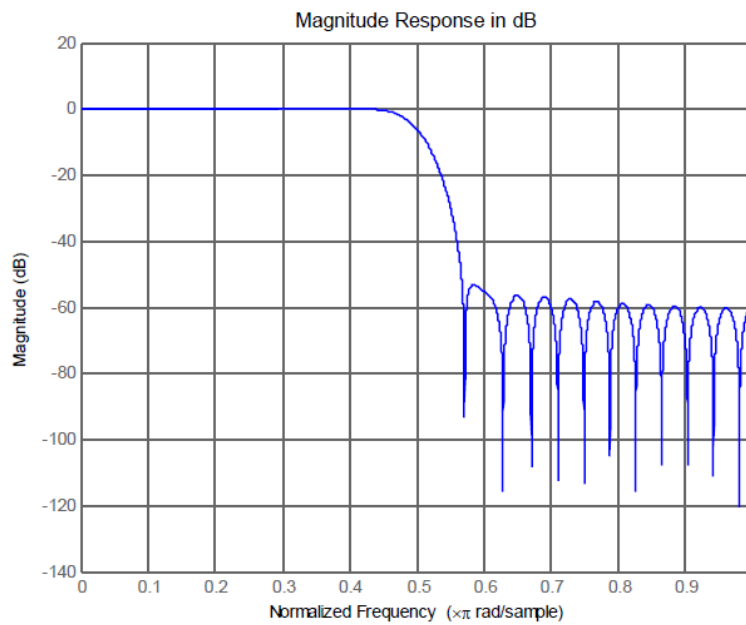
```
n=50;
```

```
wn1=0.4;
```

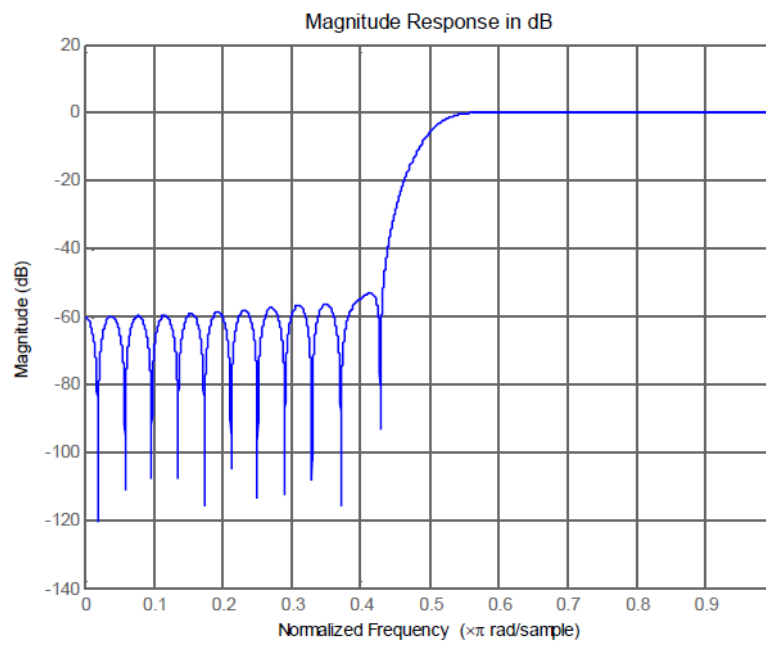
```
wn2=0.8;
```

```
b=fir1(n,[wn1,wn2]);  
fvtool(b,1);  
  
% band stop filter  
  
n=50;  
wn1=0.4;  
wn2=0.8;  
b=fir1(n,[wn1,wn2],'stop');  
fvtool(b,1);
```

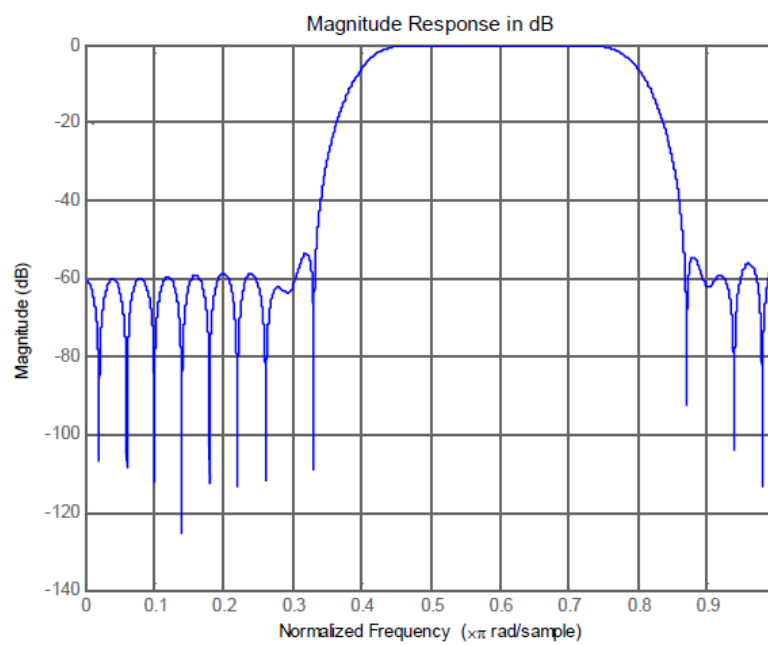
**Low pass filter:**



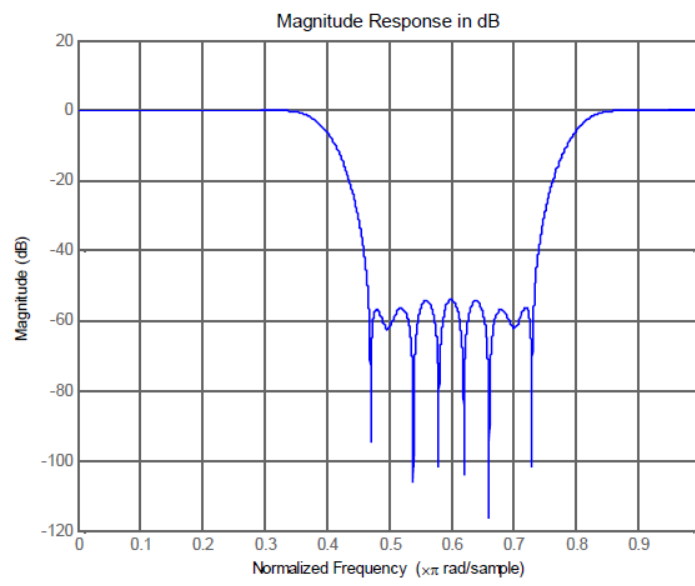
### High Pass Filter:



### Band Pass Filter:



### Band Stop Filter:





**Exp No: 9**

**Date:**

## **FIR FILTERS USING DIFFERENT WINDOWING TECHNIQUES**

### **Aim:**

To design FIR filters using different windowing techniques

### **Essentials required:**

Hardware: IBM PC or compatible

Software: MATLAB v5.1 or higher

### **Program**

%MATLAB program of FIR Low pass filter using Hanning %Hamming, Blackman and Kaiser window

```
clc;
```

```
wc=.5*pi;
```

```
N=25;
```

```
w=0:0.1:pi;
```

```
b=fir1(N,wc/pi,blackman(N+1));
```

```
h=freqz(b,1,w);
```

```
subplot(3,2,1)
```

```
plot(w/pi,abs(h))
```

```
grid;xlabel('normalised frequency');
```

```
ylabel('magnitude in dB')
```

```
title('FIR LPF USING BLACKMAN WINDOW')
```

```
b=fir1(N,wc/pi,hamming(N+1));
```

```
h=freqz(b,1,w);
```

```
subplot(3,2,2)
```

```
plot(w/pi,abs(h));
```

```
grid;
```

```
xlabel('normalised frequency');
```

```
ylabel('magnitude in dB')
```

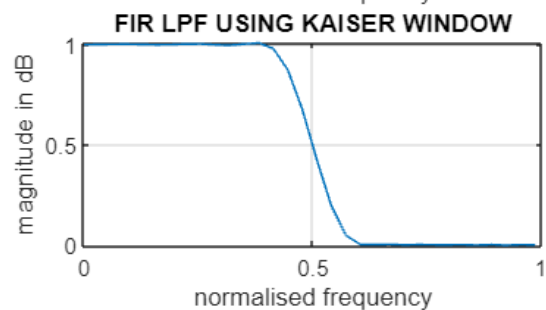
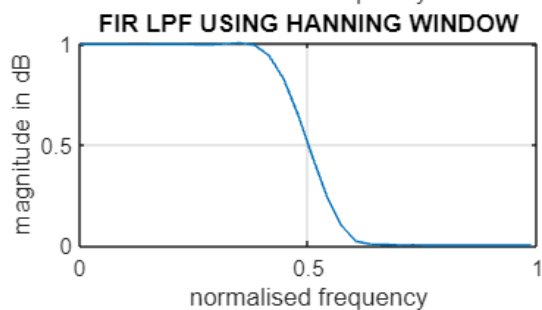
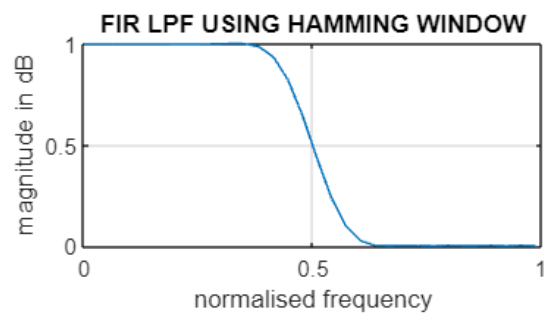
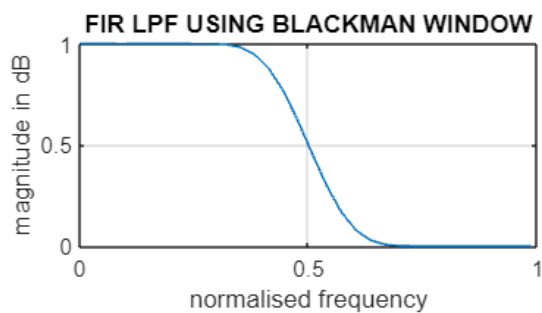
```
title('FIR LPF USING HAMMING WINDOW')
```

```

b=fir1(N,wc/pi,hanning(N+1));
h=freqz(b,1,w);
subplot(3,2,3)
plot(w/pi,abs(h));
grid;
xlabel('normalised frequency');
ylabel('magnitude in dB')
title('FIR LPF USING HANNING WINDOW')

b=fir1(N,wc/pi,kaiser(N+1,3.5));
h=freqz(b,1,w);
subplot(3,2,4)
plot(w/pi,abs(h));
grid;
xlabel('normalised frequency');
ylabel('magnitude in dB')
title('FIR LPF USING KAISER WINDOW')

```



## Result:

The MATLAB programs to design FIR filters using different windowing techniques are written and the results are plotted.

**Exp No: 10**

**Date:**

## **EFFECT OF QUANTIZATION**

**Aim:**

To design Effect of Quantization

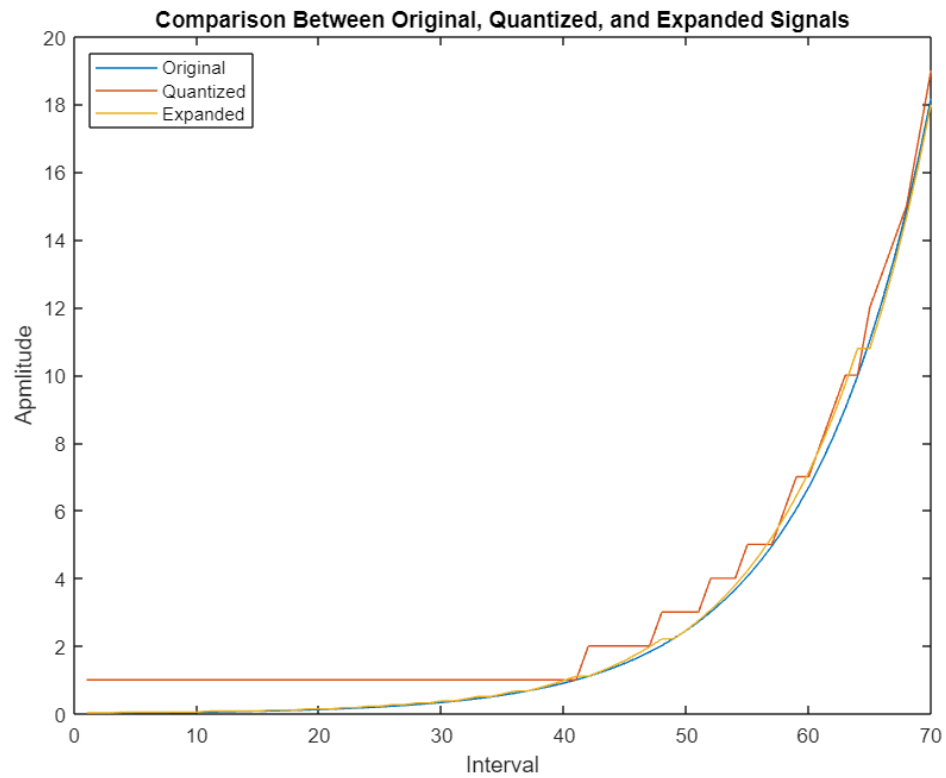
**Essentials required:**

Hardware: IBM PC or compatible

Software: MATLAB v5.1 or higher

**Program:**

```
sig = exp(-4:0.1:4);
V = max(sig);
partition = 0:2^6 - 1;
codebook = 0:2^6;
[~,qsig,distortion] = quantiz(sig,partition,codebook);
mu = 255; % mu-law parameter
csig_compressed = compand(sig,mu,V,'mu/compressor');
[~,quants] = quantiz(csig_compressed,partition,codebook);
csig_expanded = compand(quants,mu,max(quants),'mu/expander');
distortion2 = sum((csig_expanded - sig).^2)/length(sig);
[distortion, distortion2]
plot([sig' qsig' csig_expanded']);
title('Comparison Between Original, Quantized, and Expanded Signals');
xlabel('Interval');
ylabel('Amplitude');
legend('Original','Quantized','Expanded','location','nw');
axis([0 70 0 20])
```



### Result:

The MATLAB program to view the effects of Quantization is written and the results are plotted.