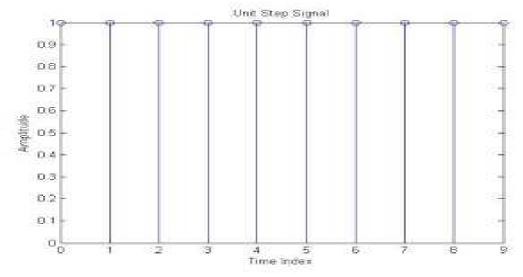
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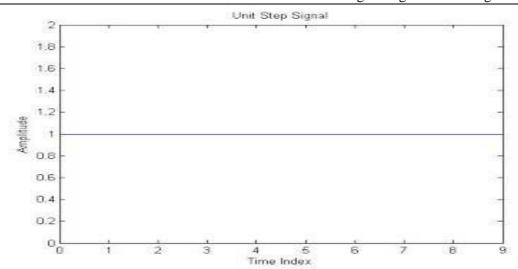
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Aim: Write a MATLAB program to generate the following elementary signals:

- iv) Unit step sequence
 iv) Sine signal
 v) Cosine signal
 vi) Unit ramp
 vi) Exponential signal
- (i) MATLAB program to generate Unit Step sequence.

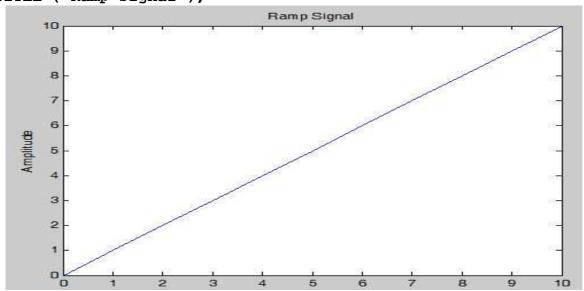


(ii) MATLAB program to generate Unit Step signal.



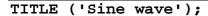
(iii) MATLAB program to generate Unit ramp Signal.

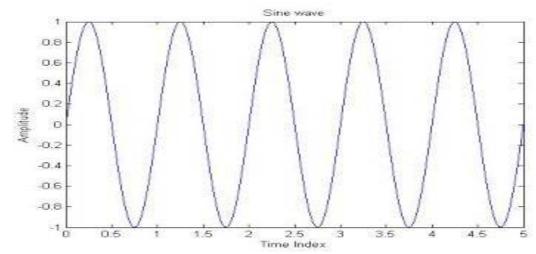
```
n=input ('Enter the length of the sequence N= ');
t=0:n;
y=t;
plot(t,y);
ylabel ('Amplitude');
xlabel ('Time Index');
TITLE ('Ramp Signal');
```



(iv) MATLAB program to generate Sine Wave Signal.

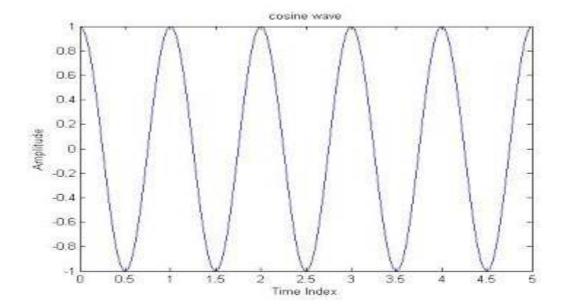
```
f= input('enter the frequency in hertz of the sine wave');
t=0:.0001:5;
y=sin(2*pi*f*t);
plot(t,y);
ylabel ('Amplitude');
xlabel ('Time Index');
```





(v) MATLAB program to generate Cosine Wave Signal.

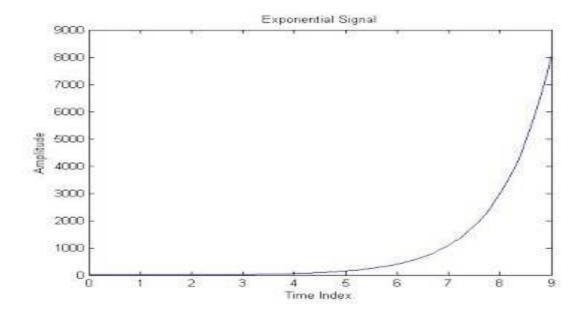
```
f= input('enter the frequency in hertz of the sine wave');
t=0:.0001:5;
y=cos(2*pi*f*t);
plot(t,y);
ylabel ('Amplitude');
xlabel ('Time Index');
TITLE ('cosine wave');
```



(vi) MATLAB program to generate Exponential signal Signal.

```
n=input('Enter the duration of the signal N = ');
a=input ('Enter the scaling factor a = ');
t=0:.1:n-1;
y=exp(a*t);
```

```
plot(t,y);
ylabel ('Amplitude');
xlabel ('Time Index');
TITLE ('Exponential Signal');
```



Date:	
Date.	

Aim: Write a MATLAB program to generate the following sequences.

a)
$$0.5*n*cos(2*\Pi*n+(\Pi/4))$$

b)
$$x1=\cos((2*\prod*n)/256)$$

$$x2 = \cos((6*\prod *n)/256)$$

$$x=x1+x2$$

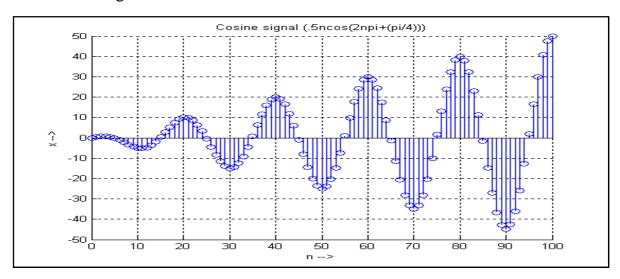
MATLAB Program.

```
% Matlab program <cosine_signal.m>
```

% This program is for generating the given cosine signal

```
clc; %Clear the window close all; %Close all files clear all; %Clear the screen m=0:1:100; %Range of n y=cos(.05*((2*pi*n)+(pi/4))); x=.5.*n.*y stem(n,x); grid on; title('Cosine signal (.5ncos(2npi+(pi/4)))'); xlabel('n -->'); ylabel('x -->');
```

MATLAB Figure:



MATLAB Program.

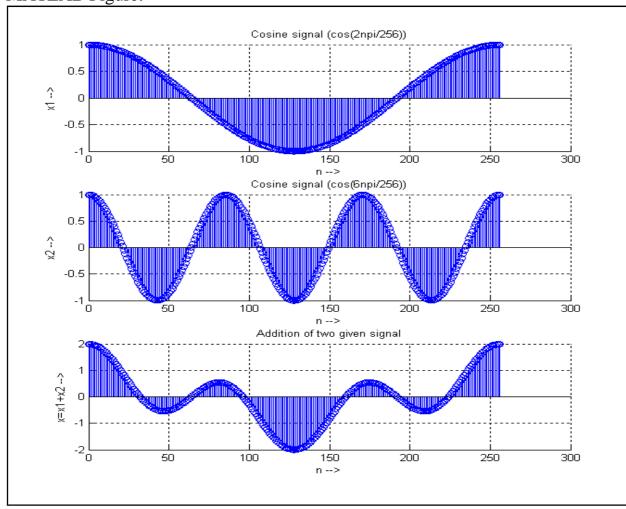
```
% Matlab program <addition_of_signal.m>
```

% This program is for adding the given two cosine signal

```
clc;
                                                              %Clear the window
                                                              %Close all files
close all;
clear all;
                                                              %Clear the screen
n=0:1:256
                                                              %Range of n
x1=\cos(2*pi*n/256);
subplot(3,1,1);
stem(n,x1);
grid on;
title ('Cosine signal (cos(2npi/256))');
xlabel('n -->');
ylabel('x1 -->');
x2 = \cos(6*pi*n/256);
subplot(3,1,2);
stem(n,x2);
grid on;
title ('Cosine signal (cos(6npi/256))');
xlabel('n -->');
ylabel('x2 -->');
```

```
x=x1+x2;
subplot(3,1,3);
stem(n,x);
grid on;
title ('Addition of two given signal');
xlabel('n -->');
ylabel('x = x1+x2 -->');
```

MATLAB Figure:



CONCLUSION:

Date:	

Aim: Write a MATLAB program for following:

- a) To illustrate the effect of up-sampling in frequency domain.
- b) To illustration of Interpolation process.

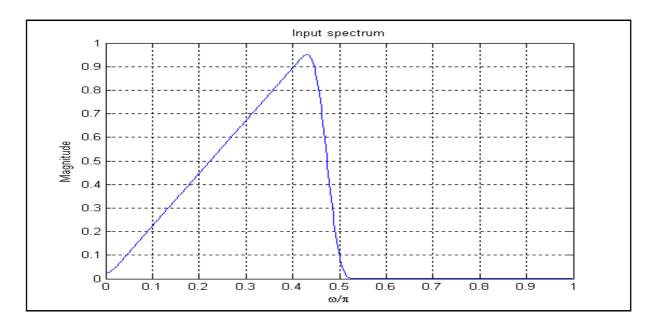
```
% Matlab program <Effect_upsampling_freq.m>
% This program gives effect of up sampling in frequency domain.
                                                            %Clear the window
clc;
close all;
                                                             %Close all files
clear all;
                                                             %Clear the screen
freq = [0\ 0.45\ 0.5\ 1];
mag = [0 \ 1 \ 0 \ 0];
x = fir2(99, freq, mag);
[Xz, w] = freqz(x, 1, 512);
                                                            % Evaluate &plot input spectrum
figure(1)
plot(w/pi, abs(Xz));
grid on;
xlabel('\omega/\pi');
ylabel('Magnitude');
title('Input spectrum');
L = input('Enter the up-sampling factor: ');
                                                            % Generate up-sampled sequence
y = zeros(1, L*length(x));
y([1: L: length(y)]) = x;
[Yz, w] = freqz(y, 1, 512);
                                                            % Evaluate &plot o/p spectrum
figure(2);
plot(w/pi, abs(Yz));
grid on;
xlabel('\omega/\pi');
ylabel('Magnitude');
```

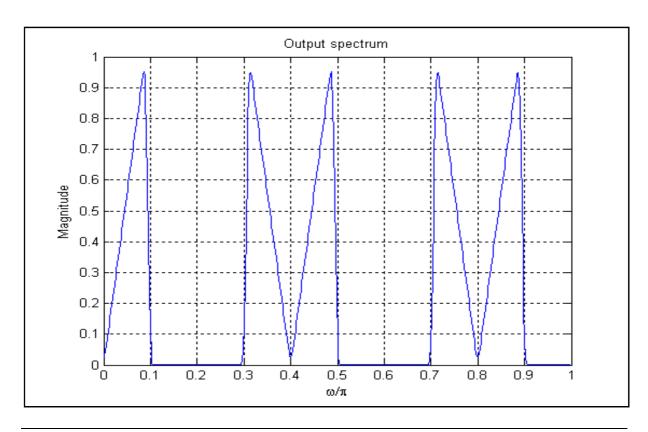
title('Output spectrum');

COMMAND WINDOW

Enter the up-sampling factor: 5

MATLAB Figures.





MATLAB Program.

```
% Matlab program <Interpolation.m>
% This program illustrate the interpolation process.
clc;
                                                            %Clear the window
close all;
                                                            %Close all files
clear all;
                                                            %Clear the screen
N = input('Length of input signal: ');
L = input('Up-sampling factor: ');
f1 = input('Frequency of first sinusoid: ');
f2 = input('Frequency of second sinusoid: ');
n = 0:N-1;
                                                            % Generate the input sequence
x = \sin(2*pi*f1*n) + \sin(2*pi*f2*n);
                                                            % Generate interpolated o/p
y = interp(x,L);
figure(1);
                                                            % Plot input & output sequences
stem(n,x(1:N));
title('Input sequence');
xlabel('Time index n');
ylabel('Amplitude');
grid on;
m=0:N*L-1;
figure(2);
stem(m,y(1:N*L));
title('Interpolated Output sequence');
xlabel('Time index n');
ylabel('Amplitude');
grid on;
```

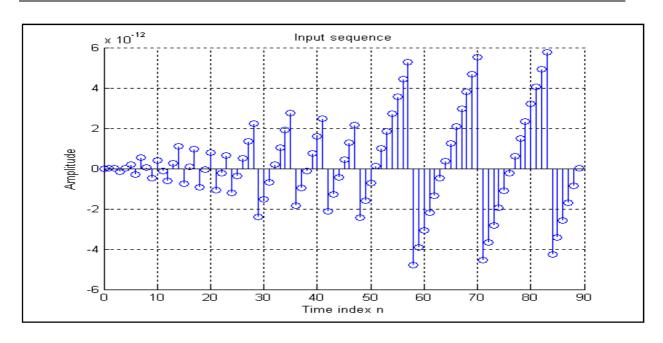
COMMAND WINDOW

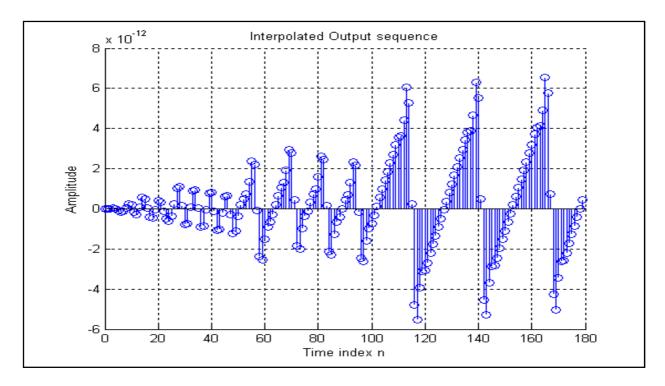
Length of input signal: 90 Up-sampling factor: 2

Frequency of first sinusoid: 50 Frequency of second sinusoid: 100

MATLAB Figure.

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CONCLUSION:

Date:	

Aim: Write a MATLAB program to illustrate Moving Average Filter for signal smoothing.

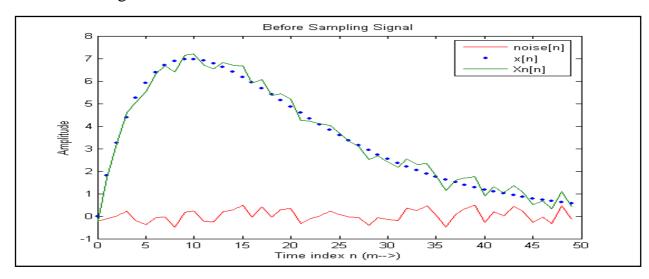
```
% Matlab program <moving_average.m>
% This program is for generating the given cosine signal
clc:
                                                            %Clear the window
close all;
                                                            %Close all files
clear all;
                                                            %Clear the screen
R = 50;
noise = rand(R,1)-0.5;
m = 0:1:R-1:
x = 2*m.*(0.9.^m);
Xn = x + noise';
figure(1)
plot(m,noise,'r-',m,x,'.',m,Xn);
title ('Before Sampling Signal');
xlabel('Time index n (m-->)');
ylabel('Amplitude');
legend('noise[n]','x[n]','Xn[n]');
M = input('Number of input samples = ');
b = ones(M,1)/M;
y = filter(b, 1, Xn);
M1 = input('Number of input samples = ');
b1 = ones(M1,1)/M1;
y1 = filter(b1,1,Xn);
figure(2)
plot(m,x,'.',m,y,'--',m,y1);
title ('After Sampling Signal');
legend('x[n]','y[n]','y1[n]');
xlabel ('Time index n');
```

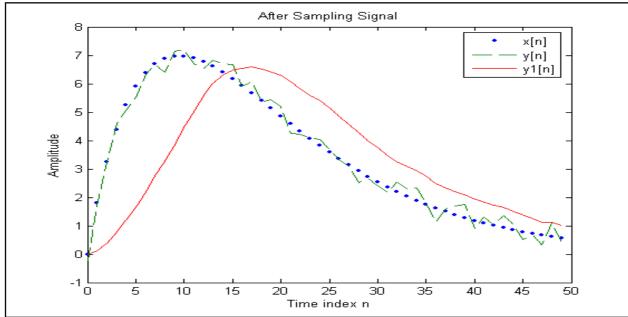
ylabel('Amplitude');

COMMAND WINDOW

Number of input samples = 1 Number of input samples = 12

MATLAB Figure:





CONCLUSION:

_	
Date:	
Date:	

Aim: Write a MATLAB program to find the linear convolution of two sequences.

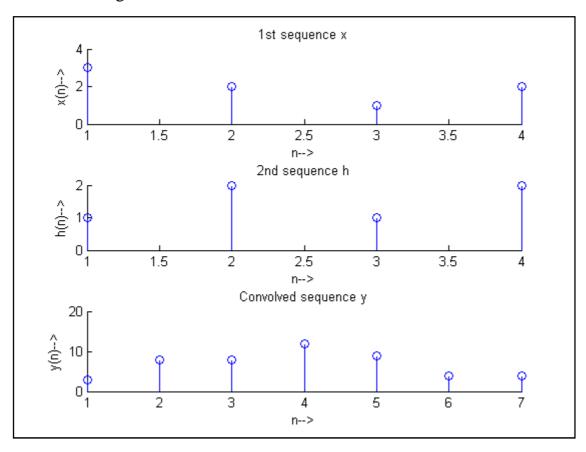
- a) Without using MATLAB convolution function.
- b) Using MATLAB convolution function.

```
% Matlab program < linear_convolution.m>
% This program performs convolution of two positive sequences.
                                                             %Clear the window
clc;
close all:
                                                             %Close all files
                                                             %Clear the screen
clear all;
                                                             %Enter the 1<sup>st</sup> sequence
x=input('Enter the 1st sequence x:');
subplot(3,1,1);
stem(x);
title('1st sequence x');
xlabel('n-->');
ylabel('x(n)-->');
                                                             %Enter the 2<sup>nd</sup> sequence
h=input('Enter the 2nd sequence h:');
subplot(3,1,2);
stem(h);
title('2nd sequence h');
xlabel('n-->');
ylabel('h(n)-->');
x1 = length(x);
                                                             %Finds the length of x
                                                             %Finds the length of h
h1=length(h);
                                                             %Gives the max. Range of n
n1 = length(x) + length(h) - 1;
for n=1:n1
                                                             %MATLAB starts any array
                                                                from index 1
  sum(n)=0;
       for k=1:x1
                                                             %Specifies summation limit
```

```
m=n+1-k;
                                                       %m=n-k
         if m \le h1
                                                       %To keep m below h1
             if k < n+1
             sum(n)=sum(n)+(x(k)*h(m));
                                                       %Convolution
             end
        else
        end
      end
  y(n)=sum(n);
end
disp ('The convolution of x(n) h(n) is y(n) = x(n) *h(n)');
                                                       %Convolved output
y
subplot(3,1,3);
stem(y);
title('Convolved sequence y');
xlabel('n-->');
ylabel('y(n)-->');
COMMAND WINDOW
```

```
Enter the sequence x:[3\ 2\ 1\ 2]
Enter the sequence h:[1\ 2\ 1\ 2]
The convolution of x(n)& h(n) is y(n)=x(n)*h(n)
y=3 8 8 12 9 4 4
```

MATLAB Figures:



- % Matlab program ear_convolution.m>
- % This program performs convolution of two positive sequences by MATLAB function.

```
clc;
                                                                 %Clear the window
close all;
                                                                 %Close all files
clear all;
                                                                 %Clear the screen
                                                                 %Enter the 1<sup>st</sup> sequence
x=input('Enter the 1st sequence x:');
subplot(3,1,1);
stem(x);
title('1st sequence x');
xlabel('n-->');
ylabel('x(n)-->');
                                                                 %Enter the 2<sup>nd</sup> sequence
h=input('Enter the 2nd sequence h:');
subplot(3,1,2);
stem(h);
title('2nd sequence h');
```

```
Enter the sequence x:[3\ 2\ 1\ 2]
Enter the sequence h:[1\ 2\ 1\ 2]
The convolution of x(n)\& h(n) is y(n) = x(n)*h(n)
y = 3 \ 8 \ 8 \ 12 \ 9 \ 4 \ 4
```

CONCLUSION:

Dotos	
Date:	

Aim: Write a MATLAB program for following:

- a) To obtain partial fraction expansion of rational Z-transform.
- b) To obtain Z-transform from partial fraction expansion.
- c) To obtain power series expansion of Z-transform.
- d) Stability test for Z-transform.

MATLAB Program.

```
% Matlab program <PFE_of_Ztrans.m>
% This program gi.ves partial fraction expansion of Z-transform.
clc;
                                                             %Clear the window
                                                             %Close all files
close all;
clear all;
                                                             %Clear the screen
num = input('Enter the numerator coefficients: ');
                                                             \% \text{ num}=(1+(1/z))
den = input('Enter the denominator coefficients: ');
                                                             % den = (1-(5/6)(1/z)+(1/6)(1/z^2))
[r,p,k] = residuez(num,den);
disp('Residues');
disp(r')
disp('Poles');
disp(p')
disp('Constants');
disp(k)
```

COMMAND WINDOW

Enter the numerator coefficients: [1 1]

Enter the denominator coefficients: [1 -5/6 1/6]

```
Residues
  9.0000 -8.0000
Poles
  0.5000 0.3333
Constants []
MATLAB Program.
% Matlab program <Ztrans_from_PFE.m>
% This program gives Z-transformer from partial fraction expansion.
                                                         %Clear the window
clc:
                                                         %Close all files
close all;
clear all;
                                                         %Clear the screen
r = input('Enter the residues: ');
p = input('Enter the poles: ');
k = input('Enter the constants: ');
[num, den] = residuez(r,p,k);
disp('Numerator polynomial coefficients');
disp(num)
disp('Denominator polynomial coefficients');
disp(den)
COMMAND WINDOW
Enter the residues: [9 -8]
Enter the poles: [.5 .3333]
Enter the constants: []
Numerator polynomial coefficients
        1.0000 1.0003
Denominator polynomial coefficients
```

1.0000 -0.8333 0.1666

MATLAB Program.

```
% Matlab program <PSE_of_ztrans.m>
```

% This program gives power series expansion of Z-transform.

clc; %Clear the window close all; %Close all files clear all; %Clear the screen

x = [1 zeros(1, N-1)]; % Compute the desired number of inverse transform coefficients

y = filter(num, den, x);
disp('Coefficients of the power series expansion');

disp(y)

COMMAND WINDOW

Enter the length of output vector: 4

Enter the numerator coefficients: [1]

Enter the denominator coefficients: [1 -2]

Coefficients of the power series expansion

1 2 4 8

MATLAB Program.

% Matlab program <Stability_test.m>

% This program is used for stabilities test of Z-transform.

clc; %Clear the window close all; %Close all files clear all; %Clear the screen

```
num = input('Enter the numerator vector: ');
den = input('Enter the denominator vector: ');
N = max(length(num), length(den)); \\
x = 1;
y0 = 0;
S = 0;
zi = zeros(1,N-1);
for n = 1:1000
  [y,zf] = filter(num,den,x,zi);
  if abs(y) < 0.000001,
       break,
  end
  x = 0;
  S = S + abs(y);
  y0 = y; zi = zf;
end
if n < 1000
  disp('Stable Transfer Function');
else
  disp('Unstable Transfer Function');
end
```

Enter the numerator vector: [1 1]

Enter the denominator vector: [1 -5/6 1/6]

Stable Transfer Function

CONCLUSION:

Dotos	
Date:	

Aim: Write a MATLAB program to:

- a) Obtain N-point DFT of a sequence.
- b) Obtain N-point IDFT of a sequence.
- c) Compute linear convolution using DFT.

```
% Matlab program <DFT.m>
% This program performs DFT of the entered sequence.
                                                          %Clear the window
clc;
                                                          %Close all files
close all;
clear all;
                                                          %Clear the screen
N=input('Enter the N-point to be calculated for DFT: ');
x=input('Enter the sequence: ');
for k=2:1:N+1
sum(k-1)=0;
   for n=1:1:N
      y(n)=x(n)*exp(-(j*2*pi*(n-1)*(k-2)/N));
      sum(k-1)=sum(k-1)+y(n);
   end
end
z=sum;
fprintf('The %d point DFT of the given input sequence is ', N)
Z
```

```
Enter the N-point to be calculated for DFT: 4
Enter the sequence: [0 1 2 3]
The 4 point DFT of the given input sequence is
z =
  6.0000
               -2.0000 + 2.0000i
                                     -2.0000 - 0.0000i
                                                            -2.0000 - 2.0000i
MATLAB Program.
% Matlab program <IDFT.m>
% This program performs IDFT of the entered sequence.
clc;
                                                          %Clear the window
                                                          %Close all files
close all;
clear all;
                                                          %Clear the screen
N=input('Enter the N-point to be calculated for DFT: ');
x=input('Enter the sequence: ');
for k=2:1:N+1
sum(k-1)=0;
   for n=1:1:N
      y(n)=x(n)*exp((j*2*pi*(n-1)*(k-2)/N));
      sum(k-1)=sum(k-1)+y(n);
   end
end
out=(1/N)*sum;
z=out;
fprintf ('The %d point DFT of the given input sequence is ', N)
\mathbf{Z}
```

```
Enter the N-point to be calculated for DFT: 4
Enter the sequence: [2 1+i 0 1-i]
The 4 point DFT of the given input sequence is
z =
               -0.0000 + 0.0000i 0.0000 + 0.0000i 1.0000 - 0.0000i
  1.0000
MATLAB Program.
% Matlab program < linear_convolution_DFT.m>
% This program performs linear convolution of the entered sequence using DFT.
                                                         %Clear the window
clc;
close all;
                                                         %Close all files
                                                         %Clear the screen
clear all;
x=input('Enter the sequence x: ');
h=input('Enter the sequence h: ');
xa=[x zeros(1,length(h)-1)];
ha=[h zeros(1,length(x)-1)];
N=length(xa);
for k1=2:1:(N+1)
   sum1(k1-1)=0;
   for n1=1:1:N
       y1(n1)=xa(n1)*exp(-(j*2*pi*(n1-1)*(k1-2)/N));
       sum1(k1-1)=sum1(k1-1)+y1(n1);
   end
end
Xk=sum1;
for k2=2:1:(N+1)
   sum2(k2-1)=0;
  for n2=1:1:N
       y2(n2)=ha(n2)*exp(-(j*2*pi*(n2-1)*(k2-2)/N));
       sum2(k2-1)=sum2(k2-1)+y2(n2);
   end
end
```

```
Hk=sum2;
Yk = Xk.*Hk;
M=length(Yk);
for k=2:1:(M+1)
   sum(k-1)=0;
   for n=1:1:M
      y(n)=Yk(n)*exp((j*2*pi*(n-1)*(k-2)/M));
      sum(k-1)=sum(k-1)+y(n);
   end
end
Y=(1/M)*sum;
disp('The Linear Convolution by DFT of given sequences is:')
Y
COMMAND WINDOW
Enter the sequence x: [1 2 2 1]
Enter the sequence h: [1 2 3]
The Linear Convolution by DFT of given sequences is:
```

9.0000 - 0.0000i 11.0000 - 0.0000i

CONCLUSION:

Y = 1.0000 - 0.0000i

8.0000 + 0.0000i

4.0000 - 0.0000i

3.0000 - 0.0000i

Date:	

Aim: Write a MATLAB program to design following filters.

- a) To design Butterworth Low Pass Filter
- b) To design Butterworth High Pass Filter
- c) To design Butterworth Band Pass Filter
- d) To design Butterworth Band Reject Filter

MATLAB Program.

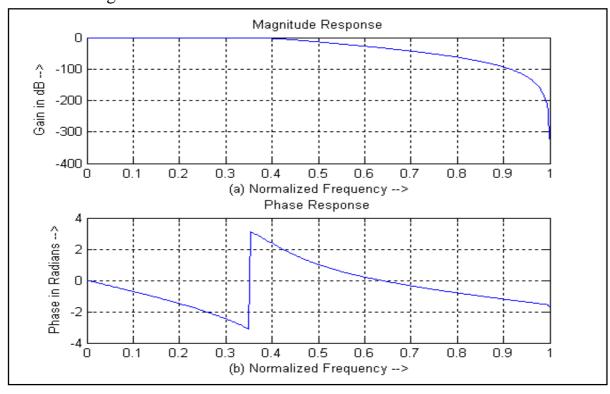
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```
% Matlab program <butter_lpf.m>
% This program gives Butterworth low pass filter.
                                                           %Clear the window
clc:
                                                           %Close all files
close all;
clear all;
                                                           %Clear the screen
Ap=input('Enter the passband attenuation in dB: ');
As=input('Enter the stopband attenuation in dB: ');
f1=input('Enter the Passband digital frequency(between 0 to 1): ');
f2=input('Enter the Stopband digital frequency(between 0 to 1): ');
fs=1;
E=((10^{(.1*Ap))-1});
L=((10^{(.1*As))-1});
N=ceil((0.5)*(log10(L/E))/log10(f2/f1))
                                                           % to find order
fc=f1/(E^{(1/(2*N))})
[b,a]=butter(N,fc)
w=0:.01:pi;
[H,NF]=freqz(b,a,w);
M=20*log10(abs(H));
                                                            %digital
subplot(2,1,1)
plot((NF/pi),M)
grid on;
title('Magnitude Response');
```

```
xlabel(' (a) Normalized Frequency -->');
ylabel(' Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
grid on;
title('Phase Response');
xlabel(' (b) Normalized Frequency -->');
ylabel(' Phase in Radians -->');
```

Enter the passband attenuation in dB: 1 Enter the stopband attenuation in dB: 15 Enter the first digital normalised freq.(in between 0 to 1): .35 Enter the second digital normalised freq.(in between 0 to 1): .6

MATLAB figure:



```
% Matlab program <butter_hpf.m>
% This program gives Butterworth high pass filter.
                                                            %Clear the window
clc;
close all;
                                                            %Close all files
clear all;
                                                            %Clear the screen
Ap=input('Enter the passband attenuation in dB: ');
As=input('Enter the stopband attenuation in dB: ');
f1=input('Enter the Passband digital frequency(between 0 to 1): ');
f2=input('Enter the Stopband digital frequency(between 0 to 1): ');
fs=1;
E=((10^{(.1*Ap))-1});
L=((10^{(1*As)})-1);
N=ceil((0.5)*(log10(L/E))/log10(f2/f1))
                                                            % to find order
fc=f1/(E^{(1/(2*N))})
[b,a]=butter(N,fc,'high')
w=0:.01:pi;
[H,NF]=freqz(b,a,w);
M=20*log10(abs(H));
                                                           %digital
subplot(2,1,1)
plot((NF/pi),M)
grid on;
title('Magnitude Response');
xlabel('(a) Normalized Frequency -->');
ylabel('Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
grid on;
title('Phase Response');
xlabel('(a) Normalized Frequency -->');
ylabel('Phase in Radians -->');
```

Enter the passband attenuation in dB: 1

Enter the stopband attenuation in dB: 15

Enter the first digital normalised freq.(in between 0 to 1): .35

Enter the second digital normalised freq.(in between 0 to 1): .6

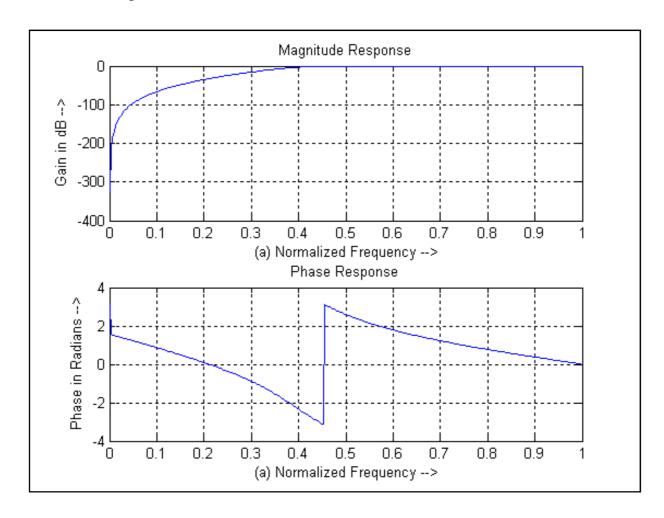
N = 5

fc = 0.4006

 $b = 0.1079 - 0.5396 \ 1.0792 - 1.0792 \ 0.5396 - 0.1079$

a = 1.0000 -0.9790 0.9695 -0.3832 0.1104 -0.0112

MATLAB figure:



```
% Matlab program <butter_bpf.m>
% This program gives Butterworth band pass filter.
                                                            %Clear the window
clc:
                                                            %Close all files
close all;
clear all;
                                                            %Clear the screen
Ap=input('Enter the passband attenuation in dB: ');
As=input('Enter the stopband attenuation in dB: ');
f1=input('Enter the first digital normalised freq.(in between 0 to 1): ');
f2=input('Enter the second digital normalised freq.(in between 0 to 1): ');
fs=1;
E=((10^{(.1*Ap))-1});
L=((10^{(1*As)})-1);
N=ceil((0.5)*(log10(L/E))/log10(f2/f1))
                                                          % to find order
fc=[f1 f2]
[b,a]=butter(N,fc,'bandpass')
w=0:.01:pi;
[H,NF]=freqz(b,a,w);
                                                          %digital
M=20*log10(abs(H));
subplot(2,1,1)
plot((NF/pi),M)
grid on;
title('Magnitude Response');
xlabel('(a) Normalized Frequency -->');
ylabel('Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
grid on;
title('Phase Response');
xlabel('(b) Normalized Frequency -->');
ylabel('Phase in Radians -->');
```

Enter the passband attenuation in dB: 1

Enter the stopband attenuation in dB: 15

Enter the first digital normalised freq.(in between 0 to 1): .35

Enter the second digital normalised freq.(in between 0 to 1): .6

N = 5

 $fc = 0.3500 \quad 0.6000$

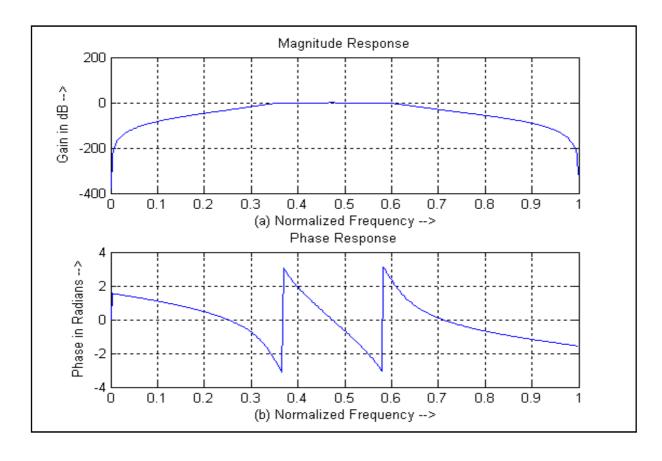
b = Columns 1 through 11

a = Columns 1 through 11

1.0000 -0.6348 2.6382 -1.3394 3.0780 -1.1743 1.8627 -0.4816 0.5776

-0.0769 0.0723

MATLAB Figure



```
% Matlab program <butter_bsf.m>
% This program gives Butterworth band stop filter.
                                                            %Clear the window
clc:
                                                            %Close all files
close all;
clear all;
                                                            %Clear the screen
Ap=input('Enter the passband attenuation in dB: ');
As=input('Enter the stopband attenuation in dB: ');
f1=input('Enter the first digital normalised freq.(in between 0 to 1): ');
f2=input('Enter the second digital normalised freq.(in between 0 to 1): ');
fs=1;
E=((10^{(.1*Ap))-1});
L=((10^{(1*As)})-1);
N=ceil((0.5)*(log10(L/E))/log10(f2/f1))
                                                           % to find order
fc=[f1 f2]
[b,a]=butter(N,fc,'stop')
w=0:.01:pi;
[H1,NF1]=freqz(b,a,w);
                                                           %digital
M=20*log10(abs(H));
subplot(2,1,1)
plot((NF/pi),M)
grid on;
title('Magnitude Response');
xlabel('(a) Normalized Frequency -->');
ylabel('Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
grid on;
title('Phase Response');
xlabel('(b) Normalized Frequency -->');
ylabel('Phase in Radians -->');
```

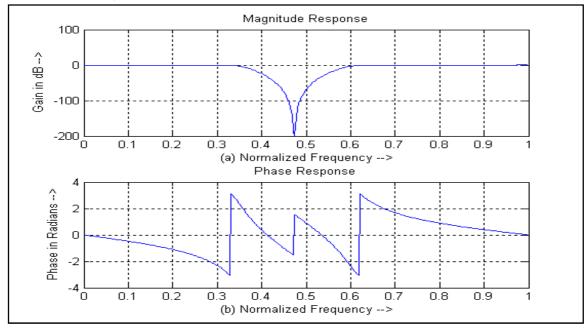
Enter the passband attenuation in dB: 1 Enter the stopband attenuation in dB: 15

Enter the first digital normalised freq.(in between 0 to 1): .35

Enter the second digital normalised freq.(in between 0 to 1): .6

MATLAB Figure:

-0.0769 0.0723



CONCLUSION:

T	
Date:	

Aim: Write a MATLAB program to design following filters.

- a) To design Chebyshev-I Low Pass Filter
- b) To design Chebyshev-I High Pass Filter
- c) To design Chebyshev-I Band Pass Filter
- d) To design Chebyshev-I Band Reject Filter

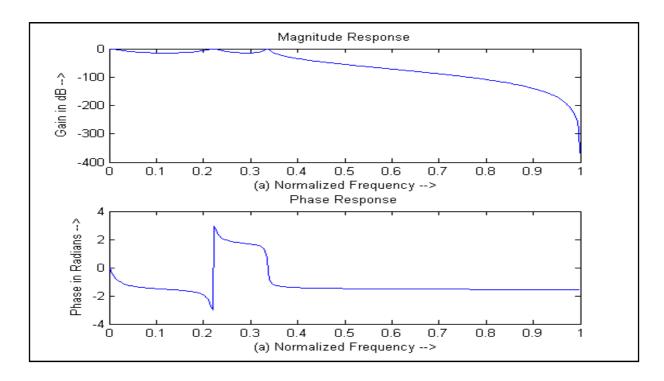
```
% Matlab program <chebyshev1_lpf.m>
% This program gives chebyshev1 low pass filter.
clc;
                                                           %Clear the window
close all;
                                                           %Close all files
clear all;
                                                           %Clear the screen
Ap=input('Enter the passband attenuation in dB: ');
As=input('Enter the stopband attenuation in dB: ');
f1=input('Enter the Passband digital frequency(between 0 to 1): ');
f2=input('Enter the Stopband digital frequency(between 0 to 1): ');
fs=1;
E=((10^{(.1*Ap))-1});
L=((10^{(.1*As))-1});
N=ceil((acosh(sqrt(L/E)))/acosh(f2/f1))
fc=f1
[b,a]=cheby1(N,Ap,fc);
w=0:.01:pi;
[H,NF]=freqz(b,a,w);
M=20*log10(abs(H));
subplot(2,1,1)
plot((NF/pi),M)
```

```
title('Magnitude Response');
xlabel(' (a) Normalized Frequency -->');
ylabel(' Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
title('Phase Response');
xlabel(' (a) Normalized Frequency -->');
ylabel(' Phase in Radians -->');
```

Enter the passband attenuation in dB: 15 Enter the stopband attenuation in dB: 50 Enter the first digital normalised freq.(in between 0 to 1): .35 Enter the second digital normalised freq.(in between 0 to 1): .6

```
\begin{array}{l} N=5\\ fc=0.3500\\ b=\!0.0006 \quad 0.0030 \quad 0.0061 \quad 0.0061 \quad 0.0030 \quad 0.0006\\ a=\!1.0000 \quad -3.4238 \quad 5.7337 \quad -5.6050 \quad 3.1947 \quad -0.8803 \end{array}
```

MATLAB figure:



```
% Matlab program <chebyshev1_hpf.m>
% This program gives Chebyshev1 high pass filter.
                                                           %Clear the window
clc:
                                                           %Close all files
close all;
clear all;
                                                           %Clear the screen
Ap=input('Enter the passband attenuation in dB: ');
As=input('Enter the stopband attenuation in dB: ');
f1=input('Enter the Passband digital frequency(between 0 to 1): ');
f2=input('Enter the Stopband digital frequency(between 0 to 1): ');
fs=1:
E=((10^{(.1*Ap))-1});
L=((10^{(.1*As))-1});
N=ceil((acosh(sqrt(L/E)))/acosh(f2/f1))
fc=f1
[b,a]=cheby1(N,Ap,fc,'High')
w=0:.01:pi;
[H,NF]=freqz(b,a,w);
M=20*log10(abs(H));
subplot(2,1,1)
plot((NF/pi),M)
title('Magnitude Response');
xlabel('(a) Normalized Frequency -->');
ylabel('Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
title('Phase Response');
xlabel('(b) Normalized Frequency -->');
ylabel('Phase in Radians -->');
```

COMMAND WINDOW

Enter the passband attenuation in dB: 15

Enter the stopband attenuation in dB: 50

Enter the first digital normalised freq.(in between 0 to 1): .35

Enter the second digital normalised freq.(in between 0 to 1): .6

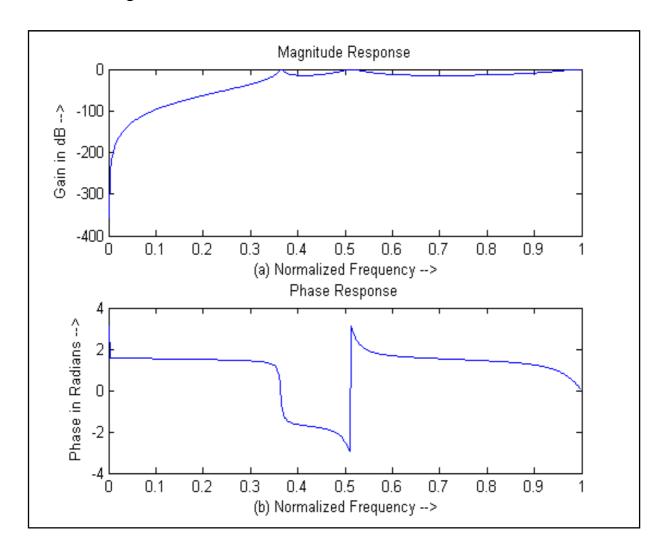
N=5

fc = 0.3500

b = 0.0177 - 0.0887 0.1774 - 0.1774 0.0887 - 0.0177

 $a = 1.0000 \quad 0.1460 \quad 1.1620 \quad 0.9530 \quad 0.2932 \quad 0.7886$

MATLAB Figure



```
% Matlab program <chebyshev1_bpf.m>
% This program gives Chebyshev1 band pass filter.
                                                            %Clear the window
clc:
                                                            %Close all files
close all;
clear all;
                                                            %Clear the screen
Ap=input('Enter the passband attenuation in dB: ');
As=input('Enter the stopband attenuation in dB: ');
f1=input('Enter the first digital normalised freq.(in between 0 to 1): ');
f2=input('Enter the second digital normalised freq.(in between 0 to 1): ');
fs=1;
E=((10^{(.1*Ap))-1});
L=((10^{(1*As)})-1);
N=ceil((acosh(sqrt(L/E)))/acosh(f2/f1))
fc=[f1 f2]
[b,a]=cheby1(N,Ap,fc,'bandpass')
w=0:.01:pi;
[H,NF]=freqz(b,a,w);
M=20*log10(abs(H));
subplot(2,1,1)
plot((NF/pi),M)
title('Magnitude Response');
xlabel('(a) Normalized Frequency -->');
ylabel('Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
title('Phase Response');
xlabel('(b) Normalized Frequency -->');
ylabel('Phase in Radians -->');
```

COMMAND WINDOW

Enter the passband attenuation in dB: 15

Enter the stopband attenuation in dB: 50

Enter the first digital normalised freq.(in between 0 to 1): .35

Enter the second digital normalised freq.(in between 0 to 1): .6

N = 5

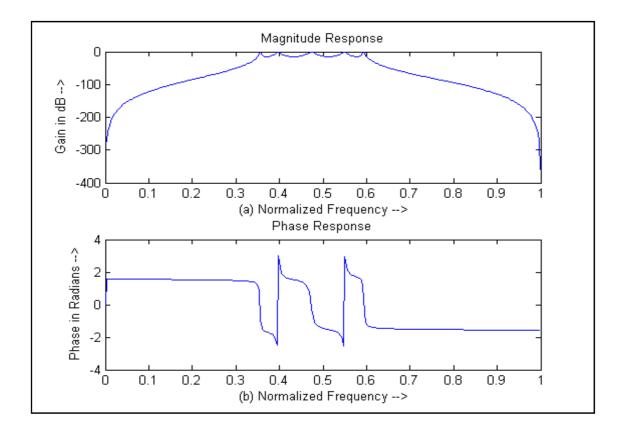
 $fc = 0.3500 \quad 0.6000$

b = Columns 1 through 11

0.0001 0 -0.0005 0 0.0011 0 -0.0011 0 0.0005 0 -0.0001

a =Columns 1 through 11

MATLAB Figure



```
% Matlab program <chebyshev1_bsf.m>
% This program gives Chebyshev1 band stop filter.
                                                           %Clear the window
clc:
                                                            %Close all files
close all;
clear all;
                                                            %Clear the screen
Ap=input('Enter the passband attenuation in dB: ');
As=input('Enter the stopband attenuation in dB: ');
f1=input('Enter the first digital normalised freq.(in between 0 to 1): ');
f2=input('Enter the second digital normalised freq.(in between 0 to 1): ');
fs=1;
E=((10^{(.1*Ap))-1});
L=((10^{(1*As)})-1);
N=ceil((acosh(sqrt(L/E)))/acosh(f2/f1))
fc=[f1 f2]
[b,a]=cheby1(N,Ap,fc,'stop')
w=0:.01:pi;
[H,NF]=freqz(b,a,w);
M=20*log10(abs(H));
subplot(2,1,1)
plot((NF/pi),M)
title('Magnitude Response');
xlabel('(a) Normalized Frequency -->');
ylabel('Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
title('Phase Response');
xlabel('(b) Normalized Frequency -->');
ylabel('Phase in Radians -->');
```

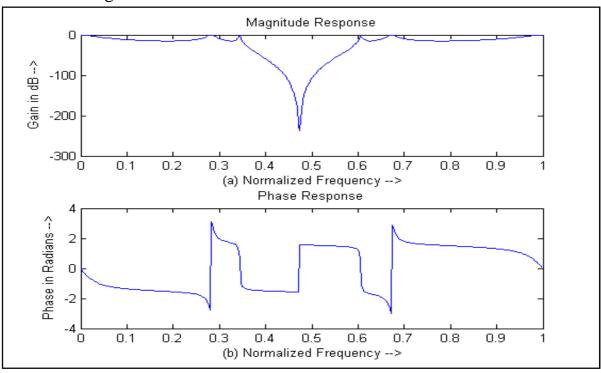
COMMAND WINDOW

Enter the passband attenuation in dB: 15 Enter the stopband attenuation in dB: 50

Enter the first digital normalised freq.(in between 0 to 1): .35 Enter the second digital normalised freq.(in between 0 to 1): .6

a = Columns 1 through 11 1.0000 -0.5229 1.2705 -0.5901 1.1767 -0.1640 -0.4844 0.3136 -0.7496 0.3767 -0.7526

MATLAB Figure:



CONCLUSION:

Practical No.: 10

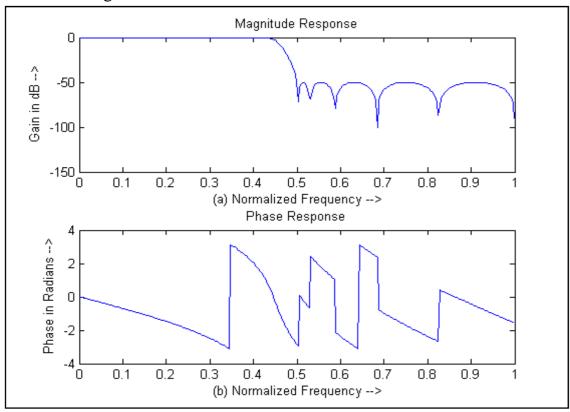
Aim: Write a MATLAB program to design following filters.

- a) To design Chebyshev-II Low Pass Filter
- b) To design Chebyshev-II High Pass Filter
- c) To design Chebyshev-II Band Pass Filter
- d) To design Chebyshev-II Band Reject Filter

```
% Matlab program <chebyshev2_lpf.m>
% This program gives chebyshev2 low pass filter.
clc;
                                                           %Clear the window
close all:
                                                           %Close all files
clear all;
                                                           %Clear the screen
Ap=input('Enter the passband attenuation in dB: ');
As=input('Enter the stopband attenuation in dB: ');
f1=input('Enter the Passband digital frequency(between 0 to 1): ');
f2=input('Enter the Stopband digital frequency(between 0 to 1): ');
fs=1;
L=((10^{(.1*As))-1});
fc=f1
N=ceil((acosh(sqrt(L)))/acosh(f2/fc))
[b,a]=cheby2(N,As,fc)
w=0:.01:pi;
[H,NF]=freqz(b,a,w);
M=20*log10(abs(H));
subplot(2,1,1)
```

```
plot((NF/pi),M)
title('Magnitude Response');
xlabel('(a) Normalized Frequency -->');
ylabel('Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
title('Phase Response');
xlabel('(b) Normalized Frequency -->');
ylabel('Phase in Radians -->');
COMMAND WINDOW
Enter the passband attenuation in dB: 10
Enter the stopband attenuation in dB: 50
Enter the first digital normalised freq.(in between 0 to 1): .5
Enter the second digital normalised freq.(in between 0 to 1): .6
fc = 0.5000
N = 11
b = Columns 1 through 12
   0.0353
          0.1607 0.4465
                            0.8844 1.3582 1.6707 1.6707 1.3582 0.8844
  0.4465 0.1607 0.0353
a = Columns 1 through 12
  1.0000 0.4636 2.1485
                            1.2185 1.8473 1.0466 0.7948 0.3697 0.1613
  0.0492 0.0108 0.0012
```

MATLAB figure:



MATLAB Program.

% Matlab program <chebyshev2_hpf.m>

% This program gives Chebyshev2 high pass filter.

clc; %Clear the window close all; %Close all files clear all; %Clear the screen

Ap=input('Enter the passband attenuation in dB: ');

As=input('Enter the stopband attenuation in dB: ');

f1=input('Enter the Passband digital frequency(between 0 to 1): ');

f2=input('Enter the Stopband digital frequency(between 0 to 1): ');

fs=1;

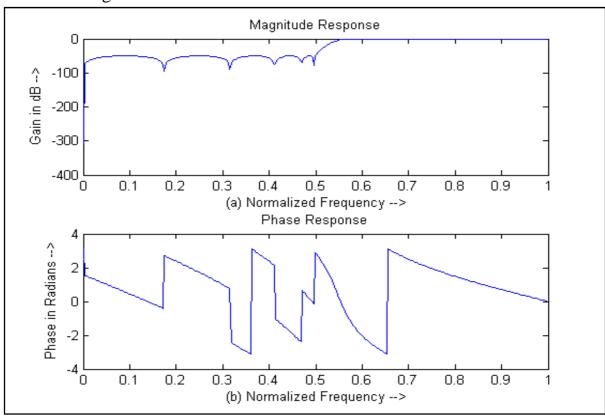
 $L=((10^{(.1*As))-1});$

fc=f1

N=ceil((acosh(sqrt(L)))/acosh(f2/fc))

[b,a]=cheby2(N,As,fc,'High')

```
w=0:.01:pi;
[H,NF]=freqz(b,a,w);
M=20*log10(abs(H));
subplot(2,1,1)
plot((NF/pi),M)
title('Magnitude Response');
xlabel('(a) Normalized Frequency -->');
ylabel('Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
title('Phase Response');
xlabel('(b) Normalized Frequency -->');
ylabel('Phase in Radians -->');
COMMAND WINDOW
Enter the passband attenuation in dB: 10
Enter the stopband attenuation in dB: 50
Enter the first digital normalised freq.(in between 0 to 1): .5
Enter the second digital normalised freq.(in between 0 to 1): .6
fc = 0.5000
N = 11
b = Columns 1 through 12
   0.0353 -0.1607 0.4465 -0.8844 1.3582 -1.6707 1.6707 -1.3582 0.8844
   -0.4465 0.1607 -0.0353
a = Columns 1 through 12
  1.0000 -0.4636 2.1485 -1.2185 1.8473 -1.0466 0.7948 -0.3697 0.1613
 -0.0492 0.0108 -0.0012
```



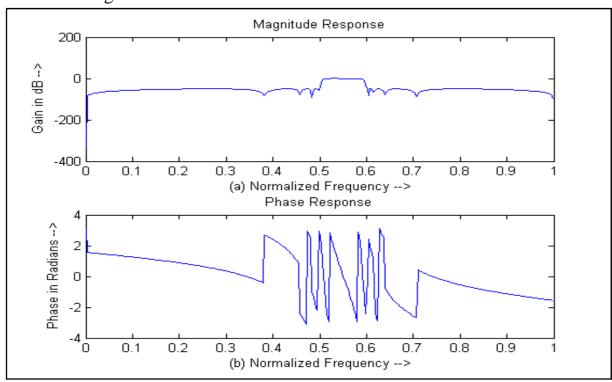
MATLAB Program.

```
% Matlab program <chebyshev2_bpf.m>
```

% This program gives Chebyshev2 band pass filter.

```
clc;
                                                             %Clear the window
                                                             %Close all files
close all;
clear all;
                                                             %Clear the screen
Ap=input('Enter the passband attenuation in dB: ');
As=input('Enter the stopband attenuation in dB: ');
f1=input('Enter the first digital normalised freq.(in between 0 to 1): ');
f2=input('Enter the second digital normalised freq.(in between 0 to 1): ');
fs=1;
L=((10^{(1*As)})-1);
N=ceil((acosh(sqrt(L)))/acosh(f2/f1))
fc=[f1 f2]
[b,a]=cheby2(N,As,fc,'Bandpass')
w=0:.01:pi;
```

```
[H,NF]=freqz(b,a,w);
M=20*log10(abs(H));
subplot(2,1,1)
plot((NF/pi),M)
title('Magnitude Response');
xlabel('(a) Normalized Frequency -->');
ylabel('Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
title('Phase Response');
xlabel('(b) Normalized Frequency -->');
ylabel('Phase in Radians -->');
COMMAND WINDOW
Enter the passband attenuation in dB: 10
Enter the stopband attenuation in dB: 50
Enter the first digital normalised freq.(in between 0 to 1): .5
Enter the second digital normalised freq.(in between 0 to 1): .6
N = 11
fc = 0.5000 \quad 0.6000
b = Columns 1 through 23
  0.0028 0.0082 0.0315 0.0629 0.1365 0.2041 0.3097 0.3517 0.3824
  -0.1365 -0.0629 -0.0315 -0.0082 -0.0028
a = Columns 1 through 23
  1.0000 3.1297 13.2170 28.7983 69.3560 115.5898 200.5175 267.4663 362.6265
  395.9562 435.1143 392.2682 354.9256 263.4227 196.7771 118.3305 72.3419
  33.9780 16.6198 5.6174 2.1010 0.4044 0.1053
```



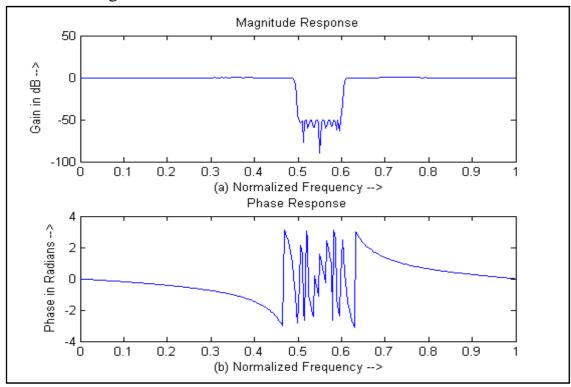
MATLAB Program.

```
% Matlab program <chebyshev2_bsf.m>
```

% This program gives Chebyshev2 band stop filter.

```
%Clear the window
clc;
close all;
                                                             %Close all files
clear all:
                                                             %Clear the screen
Ap=input('Enter the passband attenuation in dB: ');
As=input('Enter the stopband attenuation in dB: ');
f1=input('Enter the first digital normalised freq.(in between 0 to 1): ');
f2=input('Enter the second digital normalised freq.(in between 0 to 1): ');
fs=1;
L=((10^{(.1*As))-1});
N=ceil((acosh(sqrt(L)))/acosh(f2/f1))
fc=[f1 f2]
[b,a]=cheby2(N,As,fc,'stop')
w=0:.01:pi;
[H,NF]=freqz(b,a,w);
```

```
M=20*log10(abs(H));
subplot(2,1,1)
plot((NF/pi),M)
title('Magnitude Response');
xlabel(' (a) Normalized Frequency -->');
ylabel('Gain in dB -->');
an=angle(H);
subplot(2,1,2)
plot((NF/pi),an)
title('Phase Response');
xlabel('(b) Normalized Frequency -->');
ylabel(' Phase in Radians -->');LUSIO
COMMAND WINDOW
Enter the passband attenuation in dB: 10
Enter the stopband attenuation in dB: 50
Enter the first digital normalised freq.(in between 0 to 1): .5
Enter the second digital normalised freq.(in between 0 to 1): .6
N = 11
fc = 0.5000 \quad 0.6000
b = Columns 1 through 23
  0.5045 1.7364 8.1330 19.5378 52.1341 95.9104 184.2809 271.5387 407.7672
  492.1554 599.1892 597.4515 599.1892 492.1554 407.7672 271.5387 184.2809
  95.9104 52.1341 19.5378 8.1330 1.7364 0.5045
a = Columns 1 through 23
   1.0000 3.2289 14.1329 31.8420 79.5318 137.2817 247.2118 342.0046 481.8484
   546.4082 624.6946 585.6308 551.9665 426.5341 332.3406 208.3396 133.0417
   65.2213 33.3745 11.7874 4.6207 0.9302 0.2546
```



CONCLUSION:

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Date:	

Practical No.: 11

Aim: Write a MATLAB program to design FIR filter using following window.

- (a) Rectangular window
- (b) Kaiser window
- (c) Bartlett window
- (d) Blackman window
- (e) Hanning window
- (f) Hamming window

MATLAB Program.

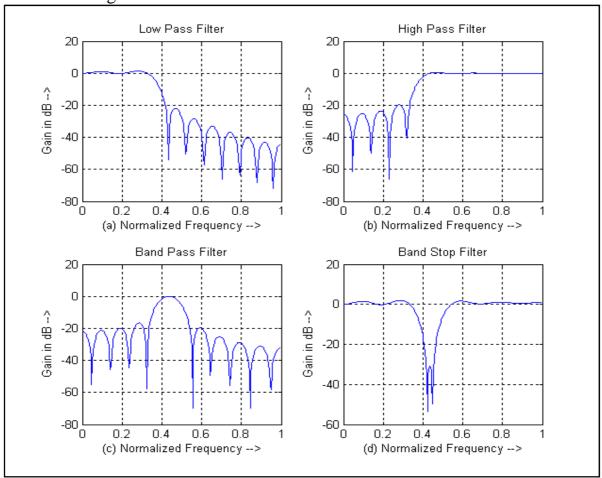
% Matlab program <FIR.m>

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```
%This program gives FIR Low pass filter, High pass filter, Band pass filter, Band stop filter.
%This program gives FIR magnitude response using different windowing techniques.
clc;
                                                            %Clear the window
close all;
                                                            %Close all files
clear all;
                                                            %Clear the screen
rp=input('Enter the pasband ripple: ');
                                                            %Enter the passband ripple
rs=input('Enter the stopband ripple: ');
                                                            %Enter the stopband ripple
fp=input('Enter the Passband frequency in hertz: ');
                                                            %Enter the passband freq.
                                                            %Enter the stopband freq.
fs=input('Enter the Stopband frequency in hertz: ');
f=input('Enter the sampling frequency in hertz: ');
                                                            %Enter the sampling freq.
wp=2*fp/f;
ws=2*fs/f;
num = -20*log10(sqrt(rp*rs))-13;
dem=14.6*(fs-fp)/f;
n=ceil(num/dem)
                                                            %To find order
n1=n+1;
if(rem(n,2) \sim = 0)
```

```
n1=n;
  n=n-1;
end
a=input('Enter the type of windowing techniques(in lower case in single quote):');
switch a
  case 'rectangular'
    y=boxcar(n1);
  case 'kaiser'
    y=kaiser(n1);
  case 'bartlett'
    y=bartlett(n1);
  case 'blackman'
    y=blackman(n1);
  case 'hamming'
    y=Hamming(n1);
  case 'hanning'
    y=Hanning(n1);
otherwise
    disp('no windowing techniques');
end
b=fir1(n,wp,y);
                                                    %Low pass filter
[h,o]=freqz(b,1,256);
m=20*log10(abs(h));
                                                    %Magnitude Response
subplot(2,2,1)
plot((o/pi),m);
title ('Low Pass Filter')
grid on;
xlabel('(a) Normalized Frequency -->');
ylabel(' Gain in dB -->');
b=fir1(n,wp,'high',y);
                                                    %High pass filter
[h,o]=freqz(b,1,256);
m=20*log10(abs(h));
subplot(2,2,2)
plot((o/pi),m);
title ('High Pass Filter')
xlabel(' (b) Normalized Frequency -->');
ylabel(' Gain in dB -->');
```

```
wn=[wp ws]
b=fir1(n,wn,y);
                                                   %Band pass filter
[h,o]=freqz(b,1,256);
m=20*log10(abs(h));
subplot(2,2,3)
plot((o/pi),m);
title ('Band Pass Filter')
grid on;
xlabel('(c) Normalized Frequency -->');
ylabel(' Gain in dB -->');
b=fir1(n,wn,'stop',y);
                                                   %Band reject filter
[h,o]=freqz(b,1,256);
m=20*log10(abs(h));
subplot(2,2,4)
plot((o/pi),m);
title ('Band Stop Filter')
grid on;
xlabel('(d) Normalized Frequency -->');
ylabel(' Gain in dB -->');
COMMAND WINDOW (for rectangular window):
Enter the passband ripple: .04
Enter the stopband ripple: .02
Enter the Passband frequency in hertz: 1500
Enter the Stopband frequency in hertz: 2000
Enter the sampling frequency in hertz: 8000
n = 20
Enter the type of windowing techniques(in lower case in single quote): 'rectangular'
wn = 0.3750 \quad 0.5000
```



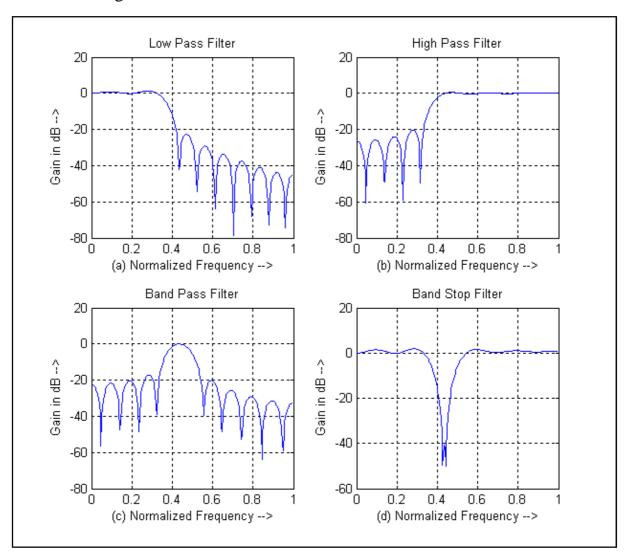
COMMAND WINDOW (for Kaiser window):

Enter the passband ripple: .04 Enter the stopband ripple: .02

Enter the Passband frequency in hertz: 1500 Enter the Stopband frequency in hertz: 2000 Enter the sampling frequency in hertz: 8000

n = 20

Enter the type of windowing techniques(in lower case in single quote): 'kaiser'



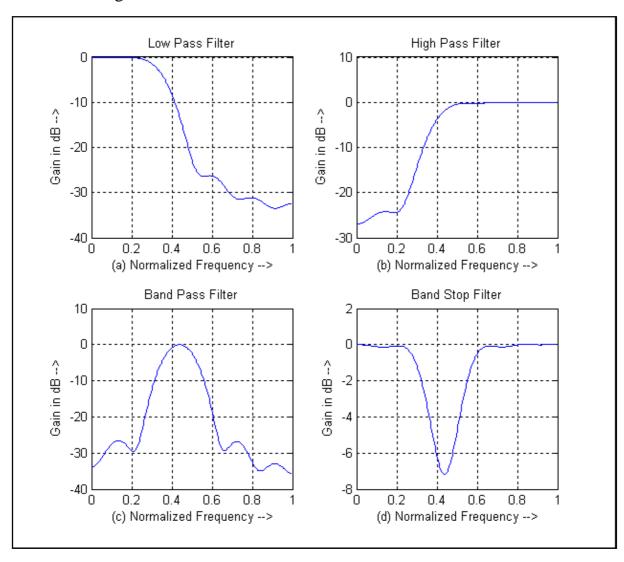
COMMAND WINDOW (for bartlett window):

Enter the passband ripple: .04 Enter the stopband ripple: .02

Enter the Passband frequency in hertz: 1500 Enter the Stopband frequency in hertz: 2000 Enter the sampling frequency in hertz: 8000

n = 20

Enter the type of windowing techniques(in lower case in single quote): 'bartlett'



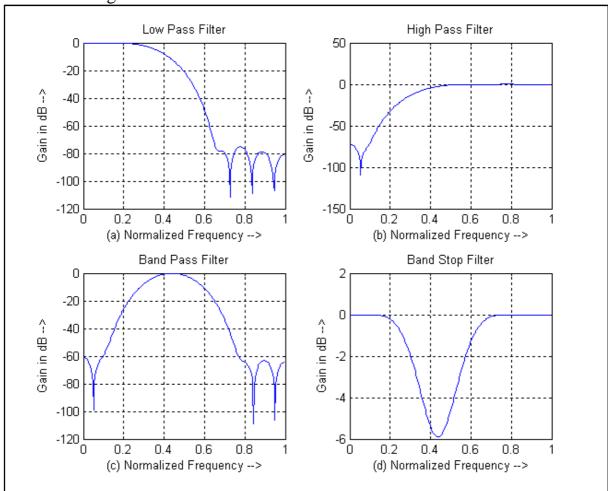
COMMAND WINDOW (for blackman window):

Enter the passband ripple: .04 Enter the stopband ripple: .02

Enter the Passband frequency in hertz: 1500 Enter the Stopband frequency in hertz: 2000 Enter the sampling frequency in hertz: 8000

n = 20

Enter the type of windowing techniques(in lower case in single quote): 'blackman'



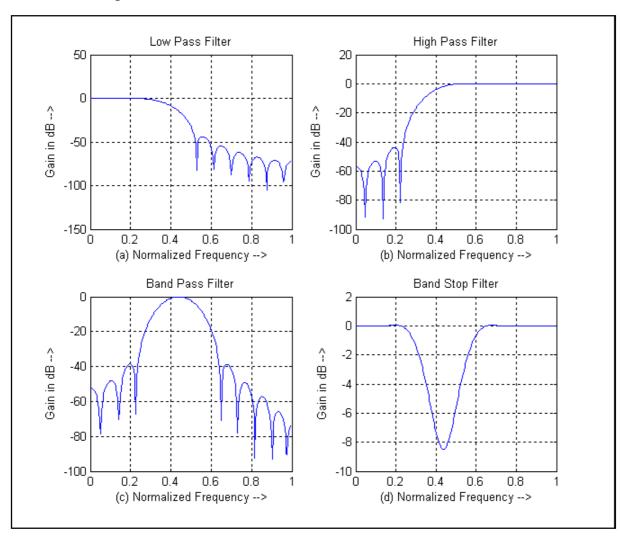
COMMAND WINDOW (for Hanning window):

Enter the passband ripple: .04 Enter the stopband ripple: .02

Enter the Passband frequency in hertz: 1500 Enter the Stopband frequency in hertz: 2000 Enter the sampling frequency in hertz: 8000

n = 20

Enter the type of windowing techniques(in lower case in single quote): 'Hanning'



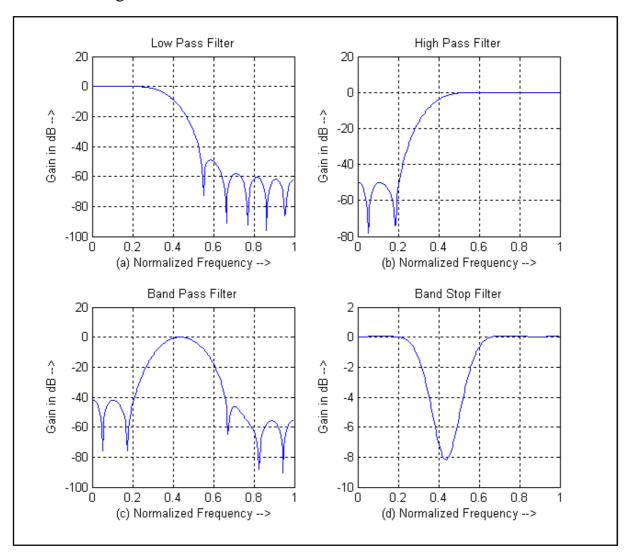
COMMAND WINDOW (for Hamming window):

Enter the passband ripple: .04 Enter the stopband ripple: .02

Enter the Passband frequency in hertz: 1500 Enter the Stopband frequency in hertz: 2000 Enter the sampling frequency in hertz: 8000

n = 20

Enter the type of windowing techniques(in lower case in single quote): 'hamming'



CONCLUSION:

Date:		
Date.		

Practical No.: 13

Aim: Write a MATLAB program to design Fast convolution method.

- a) Find linear convolution using overlap add method.
- b) Find linear convolution using overlap save method.

```
% Matlab program < Overlap_Add.m>
%This program gives convolved sequence using Overlap Add method.
                                                          %Clear the window
clc;
close all;
                                                          %Close all files
clear all;
                                                          %Clear the screen
x=input('Enter the x(n):');
                                                          % Input the sequence x(n)
h=input('Enter the h(n): ');
                                                          % Input the sequence h(n)
n1=length(x);
n2=length(h);
N=n1+n2-1;
                                                          % Length of convolved sequence
y=zeros(1,N);
h1=[h zeros(1,n2-1)];
n3=length(h1);
y=zeros(1,N+n3-n2);
H=fft(h1);
for i=1:n2:n1
                                                          % Looping for block making
  if i <= (n1+n2-1)
      x1=[x(i:i+n3-n2) zeros(1,n3-n2)];
  else
      x1=[x(i:n1) zeros(1,n3-n2)];
  end
  x2=fft(x1);
  x3=x2.*H:
  x4=round(ifft(x3));
  if (i==1)
    y(1:n3)=x4(1:n3);
  else
    y(i:i+n3-1)=y(i:i+n3-1)+x4(1:n3);
  end
end
```

```
disp('The output sequence y(n): ');
disp(y(1:N));
stem(y(1:N));
title('Overlap Add Method');
xlabel('n');
ylabel('y(n)');
```

COMMAND WINDOW

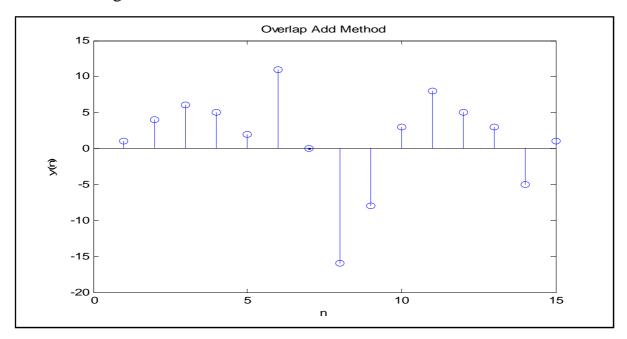
Enter the x(n): [1 2 -1 2 3 -2 -3 -1 1 1 2 -1]

Enter the h(n): [1 2 3 -1]

The output sequence y(n):

1 4 6 5 2 11 0 -16 -8 3 8 5 3 -5 1

MATLAB Figure



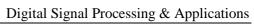
MATLAB Program.

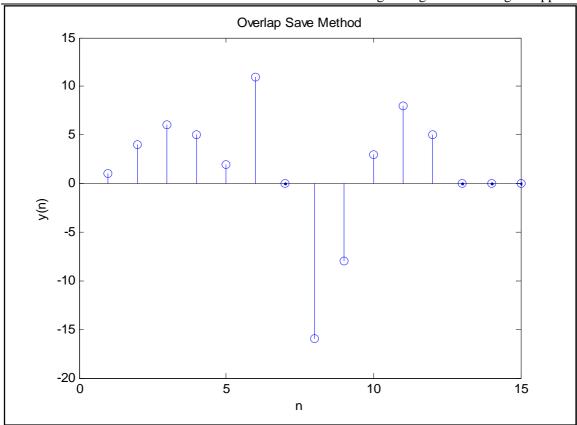
% Matlab program <Overlap_Save.m>

%This program gives convolved sequence using Overlap Save method.

clc; %Clear the window close all; %Close all files clear all; %Clear the screen

```
x=input('Enter the x(n): ');
h=input('Enter the h(n): ');
n1=length(x);
n2=length(h);
N=n1+n2-1;
h1=[h zeros(1,n2-1)];
n3=length(h1);
y=zeros(1,N);
x1=[zeros(1,n3-n2) \times zeros(1,n3)];
H=fft(h1);
for i=1:n2:n1
  y1=x1(i:i+(2*(n3-n2)));
  y2=fft(y1);
  y3=y2.*H;
  y4=round(ifft(y3));
  y(i:(i+n3-n2))=y4(n2:n3);
end
disp('The output sequence y(n): ');
disp(y(1:N));
stem(y(1:N));
title('Overlap Save Method');
xlabel('n');
ylabel('y(n)');
COMMAND WINDOW
Enter the x(n): [1 2 -1 2 3 -2 -3 -1 1 1 2 -1]
Enter the h(n): [1 2 3 -1]
The output sequence y(n):
       4 6 5 2 11 0 -16 -8 3 8 5 3 -5 1
   1
```





CONCLUSION: