

Scilab Manual for
Digital Signal Processing
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June 20, 2026

¹Funded by a grant from the National Mission on Education through ICT, <http://spoken-tutorial.org/NMEICT-Intro>. This Scilab Manual and Scilab codes written in it can be downloaded from the "Migrated Labs" section at the website <http://scilab.in>

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Experiment: 1

Analog To Digital Filter Conversion using impulse invariance method

Scilab code Solution 1.1 Analog to Digital Filter Conversion

```
1 //Filter Conversion using Impulse Invariance Method
2 //OS:Windows 10
3 //Scilab 5.5.2
4
5 clear all;
6 clc;
7 close;
8
9 s=%s;
10 T =1;
11 HS =(2)/(s^2+3*s+2) ;
12 elts = pfss(HS);
13 disp(elts, 'Factorized HS=');
14
15 //The poles associated are -2 and -1
16 p1 = -2;
17 p2 = -1;
```

```
18 z=%z;
19 HZ =(2/(1-%e^(p2*T)*z^(-1)))-(2/(1-%e^(p1*T)*z^(-1))
    )
20 disp(HZ, 'HZ=');
```

Experiment: 2

Analog To Digital Filter Conversion using bilinear transformation method

Scilab code Solution 2.2 Analog to Digital Filter Conversion

```
1 //Filter Conversion using Bilinear Transformation
  Method
2 //OS:Windows 10
3 //Scilab 5.5.2
4
5 clear all;
6 clc;
7 close;
8
9 ap = input('Enter value of ap in dB')
10 as = input('Enter value of as in dB')
11 fp = input('Enter value of fp in Hz')
12 fs = input('Enter value of fs in Hz')
13 f = input('Enter value of f')
14
15 T=1/f;
16 wp =2*%pi*fp;
```

```

17 ws =2*%pi*fs;
18 op =2/T*tan(wp*T/2);
19 os =2/T*tan(ws*T/2);
20 N= log(sqrt((10^(0.1*as)-1)/(10^(0.1*ap)-1)))/log(op
    /os);
21 disp(ceil(N), ' Order of the filter , N =');
22 s=%s;
23 HS =1/(s+1) // Transfer Function for N=1
24 oc=op //rad/sec
25 HS1 = horner(HS,oc/s);
26 disp(HS1, 'Normalized Transfer Function , H(s) =');
27 z=%z;
28 HZ= horner(HS ,(2/T)*(z-1)/(z+1));
29 disp(HZ, 'H(z)=');
30
31 //Example Input
32 //Enter value of ap in dB 3
33 //Enter value of as in dB 10
34 //Enter value of fp in Hz 1000
35 //Enter value of fs in Hz 350
36 //Enter value of f 5000
37 //Order of the filter , N = 1.
38 // Normalized Transfer Function , H(s) =
39 //      s
40 //      _____
41 //
42 //      7265.4253 + s
43 // H(z)=
44 //      1 + z
45 //      _____
46 //
47 //      -9999 + 10001z

```

Experiment: 3

Prototype Analog Low Pass Filter to High Pass Filter Conversion

Scilab code Solution 3.1 Analog LPF to HPF Conversion

```
1 // To Convert Analog LPF to High Pass
2 //Using Analog Filter Transformations
3 //For the cutoff frequency Wc = 500 Hz
4 //OS:Windows 10
5 //Scilab 5.5.2
6
7 clear all;
8 clc;
9 close;
10
11 omegap = 500;
12 omegas = 1000;
13 delta1_in_dB = -3;
14 delta2_in_dB = -40;
15 delta1 = 10^(delta1_in_dB/20)
16 delta2 = 10^(delta2_in_dB/20)
17
```

```

18 //Calculation of Filter Order
19 N = log10((1/(delta2^2))-1)/(2*log10(omegas/omegap))
20 N = ceil(N)
21 omegac = omegap;
22
23 //Poles and Gain Calculation
24 [pols ,gain]=zpbutt(N,omegac);
25 N =1;
26
27 omega_LPF = omegap; //Analog LPF Cutoff frequency
28 omega_HPF = omega_LPF; //Analog HPF Cutoff
    frequency
29 omega2 = 600; //Upper Cutoff frequency
30 omega1 = 300; //Lower Cutoff Frequency
31 omega0 = (omega2*omega1);
32 BW = omega2 - omega1; //Bandwidth
33 disp('Analog LPF Transfer function')
34 [hs,pols,zers,gain] = analpf(N,'butt',[0,0],
    omega_LPF)
35 hs_LPF = hs;
36 hs_LPF(2) = hs_LPF(2)/500;
37 hs_LPF(3)= hs_LPF(3)/500;
38 s =poly(0,'s');
39 disp('Analog HPF Transfer function')
40 h_HPF = horner(hs_LPF,omega_LPF*omega_HPF/s)
41 num = (s^2)+omega0
42 den = BW*s
43 h_BPF = horner(hs_LPF,omega_LPF*(num/den))
44
45 //Plotting Low Pass Filter Frequency Response
46 figure
47 fr=0:.1:1000;
48 hf=freq(hs_LPF(2),hs_LPF(3),%i*fr);
49 hm=abs(hf);
50 plot(fr,hm)
51 xgrid(1)
52 xtitle('Magnitude Response of LPF Filter Cutoff
    frequency = 500Hz','Analog Frequency—>','

```

```
    Magnitude ');
53
54 //Plotting High Pass Filter Frequency Response
55 figure
56 fr=0:.1:1000;
57 hf_HPF=freq(h_HPF(2),h_HPF(3),%i*fr);
58 hm_HPF=abs(hf_HPF);
59 plot(fr,hm_HPF)
60 xgrid(1)
61 xtitle('Magnitude Response of HPF Filter Cutoff
    frequency = 500Hz','Analog Frequency—>','
    Magnitude');
```

Experiment: 4

Prototype Analog Low Pass Filter to Band Pass Filter Conversion

Scilab code Solution 4.1 Analog LPF to BPF Conversion

```
1 // To Convert Analog LPF to Band Pass IIR
  Butterworth Filter
2 //Using Analog Filter Transformations
3 //For the cutoff frequency Wc = 500 Hz
4 //OS:Windows 10
5 //Scilab 5.5.2
6
7 clear all;
8 clc;
9 close;
10
11 omegap = 500;
12 omegas = 1000;
13 delta1_in_dB = -3;
14 delta2_in_dB = -40;
15 delta1 = 10^(delta1_in_dB/20)
16 delta2 = 10^(delta2_in_dB/20)
```

```

17
18 //Calculation of Filter Order
19 N = log10((1/(delta2^2))-1)/(2*log10(omegas/omegap))
20 N = ceil(N)
21 omegac = omegap;
22
23 //Poles and Gain Calculation
24 [pols,gain]=zbutt(N,omegac);
25 N =1;
26
27 omega_LPF = omegap; //Analog LPF Cutoff frequency
28 omega_HPF = omega_LPF; //Analog HPF Cutoff
    frequency
29 omega2 = 600; //Upper Cutoff frequency
30 omega1 = 300; //Lower Cutoff Frequency
31 omega0 = (omega2*omega1);
32 BW = omega2 - omega1; //Bandwidth
33 disp('Analog LPF Transfer function')
34 [hs,pols,zers,gain] = analpf(N,'butt',[0,0],
    omega_LPF)
35 hs_LPF = hs;
36 hs_LPF(2) = hs_LPF(2)/500;
37 hs_LPF(3)= hs_LPF(3)/500;
38 s =poly(0,'s');
39 h_HPF = horner(hs_LPF,omega_LPF*omega_HPF/s)
40 disp('Analog BPF Transfer function')
41 num = (s^2)+omega0
42 den = BW*s
43 h_BPF = horner(hs_LPF,omega_LPF*(num/den))
44
45 //Plotting Low Pass Filter Frequency Response
46 figure
47 fr=0:.1:1000;
48 hf=freq(hs_LPF(2),hs_LPF(3),%i*fr);
49 hm=abs(hf);
50 plot(fr,hm)
51 xgrid(1)
52 xtitle('Magnitude Response of LPF Filter Cutoff

```

```
    frequency = 500Hz', 'Analog Frequency—>', '
    Magnitude');
53
54 //Plotting Band Pass Filter Frequency Response
55 figure
56 fr=0:.1:1000;
57 hf_BPF=freq(h_BPF(2),h_BPF(3),%i*fr);
58 hm_BPF=abs(hf_BPF);
59 plot(fr,hm_BPF)
60 xgrid(1)
61 xtitle('Magnitude Response of BPF Filter Upper
    Cutoff frequency = 600Hz & Lower Cutoff frequency
    = 300Hz', 'Analog Frequency—>', 'Magnitude');
```

Experiment: 5

Prototype Analog Low Pass Filter to Band Rejection/Stop Filter Conversion

Scilab code Solution 5.1 Analog LPF to BSF Conversion

```
1 //Analog Low Pass Filter to Band Stop Filter
  Conversion
2 //OS:Windows 10
3 //Scilab 6.0.0
4
5 clear ;
6 clc ;
7 close ;
8
9 fc1 = input("Enter Analog lower cutoff freq. in Hz="
  )
10 fc2 = input("Enter Analog higher cutoff freq. in Hz="
  ")
11 fs = input("Enter Analog sampling freq. in Hz=")
12 M = input("Enter order of filter =")
13 w1 = (2*%pi)*(fc1/fs);
14 w2 = (2*%pi)*(fc2/fs);
```

```

15 disp(w1,'Digital lower cutoff frequency in radians.
    cycles/samples');
16 disp(w2,'Digital higher cutoff frequency in radians.
    cycles/samples');
17 wc1 = w1/%pi;
18 wc2 = w2/%pi;
19 disp(wc1,'Normalized digital lower cutoff frequency
    in cycles/samples');
20 disp(wc2,'Normalized digital higher cutoff frequency
    in cycles/samples');
21 [wft1,wfm1,fr1]=wfirm('lp',M+1,[wc1/2,0],'re',[0,0]);
22 disp(wft1,'Impulse Response of LPF FIR Filter:h[n]=
    ');
23
24 //Plotting the Magnitude Response of LPF FIR Filter
25 figure
26 subplot(2,1,1)
27 plot(2*fr1,wfm1)
28 xlabel('Normalized Digital Frequency  $w \longrightarrow$ ')
29 ylabel('Magnitude  $|H(w)|=$ ')
30 title('Magnitude Response of FIR LPF')
31 xgrid(1)
32 subplot(2,1,2)
33 plot(fr1*fs,wfm1)
34 xlabel('Analog Frequency in Hz  $f \longrightarrow$ ')
35 ylabel('Magnitude  $|H(w)|=$ ')
36 title('Magnitude Response of FIR LPF')
37 xgrid(1)
38
39 [wft,wfm,fr]=wfirm('sb',M+1,[wc1/2,wc2/2],'re',[0,0])
    ;
40 disp(wft,'Impulse Response of BSF FIR Filter:h[n]=')
    ;
41
42 //Plotting the Magnitude Response of HPF FIR Filter
43 figure
44 subplot(2,1,1)
45 plot(2*fr,wfm)

```

```

46 xlabel('Normalized Digital Frequency  $\omega \rightarrow$ ')
47 ylabel('Magnitude  $|H(\omega)| =$ ')
48 title('Magnitude Response of FIR BSF')
49 xgrid(1)
50 subplot(2,1,2)
51 plot(fr*fs,wfm)
52 xlabel('Analog Frequency in Hz f  $\rightarrow$ ')
53 ylabel('Magnitude  $|H(\omega)| =$ ')
54 title('Magnitude Response of FIR BSF')
55 xgrid(1)
56
57
58 //Example Input
59 //Enter Analog lower cutoff freq. in Hz=250
60 //Enter Analog higher cutoff freq. in Hz=600
61 //Enter Analog sampling freq. in Hz=2000
62 //Enter order of filter =4
63 //Digital lower cutoff frequency in radians
64 // .cycles/samples
65 // 0.7853982
66
67 // Digital higher cutoff frequency in radian
68 // s.cycles/samples
69 // 1.8849556
70
71 // Normalized digital lower cutoff frequency
72 // in cycles/samples
73 // 0.25
74
75 // Normalized digital higher cutoff frequenc
76 // y in cycles/samples
77 // 0.6
78 // Impulse Response of LPF FIR Filter:h[n]=
79 // column 1 to 3
80 // 0.1591549 0.2250791 0.25
81 // column 4 to 5
82 // 0.2250791 0.1591549
83

```

```
84 // Impulse Response of BSF FIR Filter:h[n]=
85 //      column 1 to 3
86 //      0.2527039  -0.0776516   0.65
87 //      column 4 to 5
88 //      -0.0776516   0.2527039
```

Experiment: 6

Comparison of FIR filter design using windowing method

Scilab code Solution 6.1 Rectangular Window

```
1 //Plot Magnitude Response of L.P.F.
2 //N=7 , fc =1000Hz , F=5000Hz
3 //FIR filter design using Windowing Method-
  Rectangular Window
4 //OS:Windows 10
5 //Scilab 6.0.0
6
7 clear ;
8 clc ;
9 close ;
10
11 N =7;
12 U =4;
13 h_Rect =window('re',N);
14 for n= -3+U :1:3+ U
15     if n ==4
16         hd(n) =0.4;
17     else
18         hd(n)=( sin (2* %pi *(n-U) /5) )/( %pi *(n-U
```

```

        ));
19     end
20     h(n)=hd(n)* h_Rect (n);
21 end
22 [hzm,fr]=frmag(h,256) ;
23 hzm_dB=20*log10(hzm)./max(hzm);
24 figure
25 xgrid(2);
26 plot(2*fr,hzm_dB)
27 a=gca();
28 xlabel('Frequency w*pi ');
29 ylabel('Magnitude in dB ');
30 title('Frequency Response of FIR LPF with N=7');
31 xgrid(2)
32 disp(h," Filter Co efficients , h(n)=");

```

Scilab code Solution 6.2 Kaiser Window

```

1 //Plot Magnitude Response of L.P.F .
2 //FIR filter design using Windowing Method– Kaiser
  Window
3 //OS:Windows 10
4 //Scilab 6.0.0
5
6 clear ;
7 clc ;
8 close ;
9
10 wsf = input("Enter the value of wsf in rad/sec");
11 ws = input("Enter the value of ws in rad/sec");
12 wp = input("Enter the value of wp in rad/sec");
13 as = input("Enter the value of as in dB");
14 ap = input("Enter the value of ap in dB");
15
16 B=ws -wp;

```

```

17 wc =0.5*(ws+wp);
18 wc1 =wc*2*%pi/ wsf ;
19 delta1 =10^(-0.05* as);
20 delta2 =(10^(0.05* as) -1)/(10^(0.05* as)+1);
21 delta = min(delta1,delta2);
22 alphas = -20*log10(delta);
23 alpha =0.5842*(alphas -21)^0.4+0.07886*(alphas -21)
24 D=(alphas -7.95)/14.36;
25 N1=wsf*D/B +1;
26 N= ceil(N1);
27 U= ceil(N/2) ;
28 win_1 = window('kr',N,alpha);
29 for n=- floor(N/2)+U :1: floor(N/2) +U
30     if n== ceil(N /2) ;
31         hd(n) =0.5;
32     else
33         hd(n)=(sin(%pi*(n-U)/2))/(%pi*(n-U));
34     end
35     h(n)=hd(n)* win_1(n);
36 end
37 [hzm,fr]=frmag(h,256) ;
38 hzm_dB = 20*log10(hzm)./max(hzm);
39 figure
40 plot(2*fr,hzm_dB )
41 a= gca();
42 xlabel('Frequency w*pi ');
43 ylabel('Magnitude in dB ');
44 title('Frequency Response of LPF');
45 xgrid(2);
46 disp(h," Filter Co efficients , h(n)=");
47
48 //Example Input
49 //Enter the value of wsf in rad/sec 100
50 //Enter the value of ws in rad/sec 30
51 //Enter the value of wp in rad/sec 20
52 //Enter the value of as in dB 44
53 //Enter the value of ap in dB 0.1
54 //Filter Co efficients , h(n)=

```

```

55 //      0.002441   -3.172D-18   -0.0068491   6.235D-18
      0.0145007   -9.826D-18   -0.027237   1.347D-17
      0.0494369   -1.661D-17   -0.0970495   1.874D-17
      0.3152014    0.5    0.3152014    1.874D-17
      -0.0970495   -1.661D-17    0.0494369    1.347D-17
      -0.027237   -9.826D-18    0.0145007    6.235D-18
      -0.0068491   -3.172D-18    0.002441

```

Scilab code Solution 6.3 Hamming Window

```

1 //Plot Magnitude Response of ideal H.P.F.
2 //wcl=0.25*pi , N=11
3 //FIR filter design using Windowing Method-Hamming
  Window
4 //OS:Windows 10
5 //Scilab 6.0.0
6
7 clear;
8 clc;
9 close;
10
11 N =11;
12 U =6;
13 h_hamm=window('hm',N);
14 for n= -5+U :1:5+ U
15     if n ==6
16         hd(n) =0.75;
17     else
18         hd(n)=(sin(%pi*(n-U))-sin(%pi*(n-U)/4))/(%pi
                *(n-U));
19     end
20     h(n)= h_hamm(n)*hd(n);
21 end
22 [hzm,fr]=frmag(h,256);
23 hzm_dB=20*log10(hzm)./max(hzm);

```

```

24 figure
25 plot(2*fr, hzm_dB)
26 a=gca();
27 xlabel('Frequency w*pi');
28 ylabel('Magnitude in dB');
29 title('Frequency Response of FIR HPF with N=11 using
        Hamming Window');
30 xgrid(2);

```

Scilab code Solution 6.4 Hanning Window

```

1 //Plot Magnitude Response of ideal H.P.F.
2 //N=11 ,wc1=0.25* pi
3 //FIR filter design using Windowing Method–Hanning
  Window
4 //OS:Windows 10
5 //Scilab 6.0.0
6
7 clear;
8 clc;
9 close;
10
11 N =11;
12 U =6;
13 h_hann = window('hn',N);
14 for n= -5+U:1:5+U
15     if n == 6
16         hd(n) = 0.75;
17     else
18         hd(n)=(sin(%pi*(n-U))-sin(%pi*(n-U)/4))/(%pi
                *(n-U));
19     end
20     h(n)= h_hann(n)*hd(n);
21 end
22 [hzm,fr]= frmag(h,256) ;

```

```
23 hzm_dB =20*log10(hzm)./max(hzm);
24 figure
25 plot(2*fr,hzm_dB)
26 a=gca();
27 xlabel('Frequency w*pi ');
28 ylabel('Magnitude in dB ');
29 title('Frequency Response of FIR HPF with N=11 using
        Hanning Window ');
30 xgrid(2);
```

Experiment: 7

Detection of DTMF signal

Scilab code Solution 7.1 DTMF Decoding

```
1 //DTMF Signal Detection
2 //OS:Windows 10
3 //Scilab 5.5.2
4
5 clear all;
6 clc;
7 close;
8
9 row_f1=[800 870 950 990]; // Row Frequency
10 colum_f1=[1340 1440 1540]; // Column Frequency
11 fs=8000; // Sampling Frequency
12 N=1:800; // Total No. of
    Samples for each Digit
13 mobile=[5 6 7 8 9 0 1 2 3 4];
14 total_signal=[];
15
16 figure;
17
18 for i=1:length(mobile)
19     select mobile(i)
20     case 1
```

```

21         row_f=1;
22         colum_f=1;
23     case 2
24         row_f=1;
25         colum_f=2;
26     case 3
27         row_f=1;
28         colum_f=3;
29     case 4
30         row_f=2;
31         colum_f=1;
32     case 5
33         row_f=2;
34         colum_f=2;
35     case 6
36         row_f=2;
37         colum_f=3;
38     case 7
39         row_f=3;
40         colum_f=1;
41     case 8
42         row_f=3;
43         colum_f=2;
44     case 9
45         row_f=3;
46         colum_f=3;
47     case 0
48         row_f=4;
49         colum_f=2;
50     else
51         row_f=4;
52         colum_f=1;
53     end
54     y=sin(2*3.14*(row_f1(row_f)/fs)*N)+sin(2*3.14*(
        colum_f1(colum_f)/fs)*N); //Time Domain
        Signal Generation for each Digit
55     total_signal=[total_signal y zeros(1,8800)];
56     temp(:,:,i)=y(:,:,);

```

```

57     end
58     plot(total_signal);
59     title('DTMF Signal','color','blue');
60     xlabel("Samples","color","blue");
61     ylabel("Amplitude","color","blue");
62     sound(total_signal,fs);
63
64     row_f=[];
65     col_f=[];
66
67     for i=1:10
68         n=length(temp(:,:,i));
69         p=abs(fft(temp(:,:,i))); // FFT of Signal of
            respective Digit
70         f=(0:n-1)*fs/n; // Total Frequency
            Range
71         //plot(f,p);
72         row=p(2:100); // Row Frequency
            separation
73         col=p(101:200); // Column Frequency
            separation
74         [r1 c1]=find(row==max(row)); // Finding the
            location of peak for Row Frequency
75         [r2 c2]=find(col==max(col)); // Finding the
            location of peak for Column Frequency
76         row_f=[row_f 10*c1]; // Array
            containing peak of Row Frequency
77         col_f=[col_f (10*(c2+100))-10]; // Array
            containing peak of Column Frequency
78     end
79
80     mobile_find=[]; // Blank Array to Store Mobile
            Number
81     for i=1:10 // Loop for Finding the Number form
            the Row and Column Frequency
82         if(row_f(i)==800 & col_f(i)==1340)
83             n0=1;
84             elseif(row_f(i)==800 & col_f(i)==1440)

```

```
85     n0=2;
86     elseif(row_f(i)==800 & col_f(i)==1540)
87     n0=3;
88     elseif(row_f(i)==870 & col_f(i)==1340)
89     n0=4;
90     elseif(row_f(i)==870 & col_f(i)==1440)
91     n0=5;
92     elseif(row_f(i)==870 & col_f(i)==1540)
93     n0=6;
94     elseif(row_f(i)==950 & col_f(i)==1340)
95     n0=7;
96     elseif(row_f(i)==950 & col_f(i)==1440)
97     n0=8;
98     elseif(row_f(i)==950 & col_f(i)==1540)
99     n0=9;
100    elseif(row_f(i)==990 & col_f(i)==1440)
101    n0=0;
102 end
103 mobile_find=[mobile_find n0]; // Array containing
    Decoded Digit of Mobile Number.
104 end
105
106 disp(mobile_find,"Decoded Mobile Number :");
```

Experiment: 8

Implementation of Decimation

Scilab code Solution 8.1 Decimation

```
1 //Implementation of Decimation
2 //OS:Windows 10
3 //Scilab 5.5.2
4
5 clear all;
6 clc;
7 close;
8
9 t=0:0.00025:1;
10 x=sin(2*%pi*30*t)+sin(2*%pi*60*t); //original
    signal
11 y = x(1:4:length(x)); //downsampled by a factor of 4
12 subplot(2,1,1)
13 plot2d3(0:120,x(1:121),-1);
14 xgrid(1);
15 xtitle('Original singal')
16 subplot(2,1,2)
17 plot2d3(0:30,y(1:31),-1);
18 xgrid(1);
19 xtitle('Downsampled Signal');
```

Experiment: 9

Implementation of Interpolation

Scilab code Solution 9.1 Interpolation

```
1 //Implementation of Interpolation
2 //OS:Windows 10
3 //Scilab 5.5.2
4
5 clear all;
6 clc;
7 close;
8
9 t=0:0.00025:1;
10 x=sin(2*%pi*30*t)+sin(2*%pi*60*t); //original
    signal
11 upsampling_x = zeros(1,2*length(x)); //upsampled by
    a factor of 2
12 upsampling_x(1:2:2*length(x)) = x;
13 subplot(2,1,1)
14 plot2d3(0:120,x(1:121),-1);
15 xgrid(1);
16 xtitle('Original singal')
17 subplot(2,1,2)
```

```
18 plot2d3(0:250,upsampling_x(1:251),-1);
19 xgrid(1);
20 xtitle('Upsampled Signal');
```

Experiment: 10

FIR Filter using Frequency Sampling Method

Scilab code Solution 10.1 Frequency Sampling Method

```
1 //FIR LPF Filter using Frequency Sampling Method
2 //OS:Windows 10
3 //Scilab 5.5.2
4
5 clc;
6 clear;
7 close;
8
9 N = input("Enter the value of N:");
10 U = input("Enter the value of U:");
11 for n =0+ U :1: N -1+ U
12     h(n)=(1+cos(2*%pi*(7-n)/N))/N;
13 end
14 [hz,f]=frmag(h,256) ;
15 hz_dB=20*log10(hz) ./max(hz);
16 figure ;
17 plot(2*f, hz_dB);
18 a=gca();
19 xlabel('Frequency w          pi ');
```

```
20 ylabel('Magnitude in dB ');
21 title('Frequency Response of FIR LPF');
22 xgrid(2)
23
24 //Example input
25 //Enter the value of N:15
26 //Enter the value of U:1 ,  $W_c=\pi/4$ 
```
