### Scilab Textbook Companion for Digital Signal Processing: A Modern Introduction by A. Ashok<sup>1</sup>

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# **Book Description**

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Scilab numbering policy used in this document and the relation to the above book.

Exa Example (Solved example)

Eqn Equation (Particular equation of the above book)

**AP** Appendix to Example(Scilab Code that is an Appednix to a particular Example of the above book)

For example, Exa 3.51 means solved example 3.51 of this book. Sec 2.3 means a scilab code whose theory is explained in Section 2.3 of the book.

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### Chapter 2

### Discrete Signals

Scilab code Exa 2.1a Signal energy and power

```
1 //example 2.1.a,pg no.11
2 for i=1:1:50
3     x(1,i)=3*(0.5)^(i-1);
4 end
5 //summation of x
6 E=0
7 for i=1:1:50
8     E=E+x(1,i)^2;
9 end
10 disp("the energy of given signal is")
11 E
```

Scilab code Exa 2.1b Average power of periodic signals

```
1 //example 2.1b,pg.no.11
2 n=1:1:10;
3 xn=6*cos((2*%pi*n')/4);
4 a=4;
```

```
5 p=0;
6 for i=1:1:a
7    p=p+abs(xn(i)^2);
8 end
9 P=p/a;
10 disp("The average power of given signal is")
11 P
```

#### Scilab code Exa 2.1c Average power of periodic signals

```
//example 2.1c,pg.no.11
n=1:4;
xn=6*%e^((%i*%pi*n')/2);
a=4;
p=0;
for i=1:1:a
    p=p+abs(xn(i)^2);
end
P=p/a;
disp("The average power of given signal is")
P
```

#### Scilab code Exa 2.2 Operations on Discrete Signals

```
1 //example 2.2,pg no.12
2 x=[0 2 3 4 5 6 7];
3 n1=-3:1:3;
4 y=x;
5 n2=0:1:6;
6 f=x;
7 n3=-5:1:1;
```

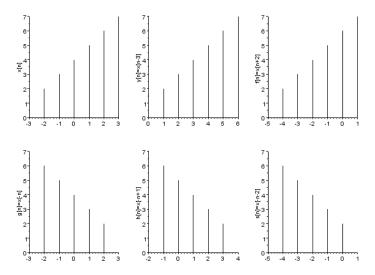


Figure 2.1: Operations on Discrete Signals

```
8 g=x(length(x):-1:1);
9 n4 = -3:1:3;
10 h=x(length(x):-1:1);
11 n5 = -2:1:4;
12 s=x(length(x):-1:1);
13 n6=-5:1:1;
14 a=gca();
15 subplot (231);
16 plot2d3('gnn',n1,x);
17 ylabel('x[n]');
18 subplot (232);
19 plot2d3('gnn',n2,y);
20 ylabel('y[n]=x[n-3]');
21 subplot (233);
22 plot2d3('gnn',n3,f);
23 ylabel('f[n]=x[n+2]');
24 subplot (234);
25 plot2d3('gnn',n4,g);
26 ylabel('g[n]=x[-n]');
27 subplot (235);
```

```
28 plot2d3('gnn',n5,h);
29 ylabel('h[n]=x[-n+1]');
30 subplot(236);
31 plot2d3('gnn',n6,s);
32 ylabel('s[n]=x[-n-2]');
```

#### Scilab code Exa 2.3a Even and Odd parts of Discrete signals

```
1 //example 2.3a pg.no.14
2 clear; clc; close;
3 n = -2:2;
4 \times 1 = [4 -2 4 -6 0];
5 x2=0.5*x1//x[n]
6 x3=0.5*[x1(length(x1):-1:1)];//x[-n]
7 xe=(x2+x3);//even part
8 \text{ xo} = (x2-x3); // \text{odd part}
9 a=gca();
10 a.thickness=2;
11 a.x_location="middle";
12 a.y_location="middle";
13 plot2d3('gnn',n,xe,rect=[-4 -6 4 6])
14 xtitle('graphical representation of even part of x[n
      ] ', 'n', 'x[n]')
15 xset ('window',1)
16 b=gca();
17 b.thickness=2;
18 b.y_location="middle";
19 b.x_location="middle";
20 plot2d3('gnn',n,xo,rect=[-2 -4 2 4])
21 xtitle ('graphical representation of odd part of x[n]
      ', 'n', 'x[n]')
```

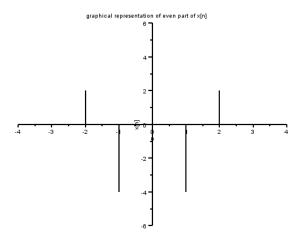


Figure 2.2: Even and Odd parts of Discrete signals

#### Scilab code Exa 2.3b Even and Odd parts of Discrete signals

```
1 //example 2.3a pg.no.14
2 clear; clc; close;
3 x1=[0 0 0 0 0 1 1 1 1 ];
4 n=-4:4;
5 x2=0.5*x1//x[n]
6 x3=0.5*[x1(length(x1):-1:1)]//x[-n]
7 xe=(x2+x3); //even part
8 xo=(x2-x3); //odd part
9 a=gca();
10 a.thickness=2;
11 a.y_location="middle";
12 a.x_location="middle";
13 plot2d3('gnn',n,xe,rect=[-4 -1 4 1]);
```

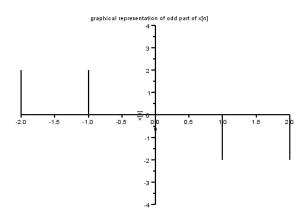


Figure 2.3: Even and Odd parts of Discrete signals

Scilab code Exa 2.4a Decimation and Interpolation of Discrete signals

```
1 //example 2.4a pg.no.17
```

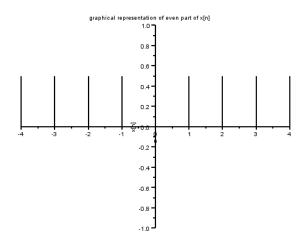


Figure 2.4: Even and Odd parts of Discrete signals

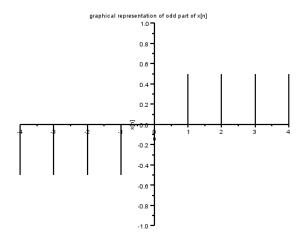


Figure 2.5: Even and Odd parts of Discrete signals

```
2 x=[1 2 5 -1];
3 xm=2;//denotes 2nd sample has pad.
4 y=[x(1:2:xm-2),x(xm:2:length(x))]//decimation
5 h=[x(1:1/3:length(x))]//step interpolated
6 g=h;
7 for i=2:3
8     g(i:3:length(g))=0;
9 end
10 //zero interpolated
11 x1=1:3:3*length(x);
12 s=interpln([x1;x],1:10)//linear interpolated
```

Scilab code Exa 2.4b Decimation and Interpolation of Discrete signals

```
1 //example 2.4 b,c.pg.no.17
2 x=[3 4 5 6];
3 xm=3;//denotes 3rd sample has pad
4 xm=xm-1;//shifting
5 g=[x(xm-2:-2:1),x(xm:2:length(x))]//decimation
6 xm=3;
7 h=[x(1:1/2:length(x))]//step interpolated
```

Scilab code Exa 2.4c Decimation and Interpolation of Discrete signals

```
//example 2.4c,pg.no.17
x=[3 4 5 6];
xm=3;
xm=xm+1*(xm-1);//shift in pad due to interpolation
xm=xm-2//normal shifting
x1=[x(1:1/3:length(x))]//step interpolated
xm=3;
xm=xm+2*(xm-1)//shift in pad due to interpolation
y=[x1(1:2:xm-2),x1(xm:2:length(x1))]//decimation
```

#### Scilab code Exa 2.4d Decimation and Interpolation of Discrete signals

```
1 //example 2.4d,pg.no.17
2 x=[2 4 6 8]
3 xm=3;//denote 3rd sample has pad
4 x1=[1 3 5 7]
5 x2=interpln([x1;x],1:6)
6 xm=xm+1*(xm-1);//shift in pad due to interpolation
7 xm=xm-1//shift in pad due to delay
8 y=[x2(2:2:xm-2),x2(xm:2:length(x2))]//decimation
```

#### Scilab code Exa 2.5 Describing Sequences and signals

```
1 //example 2.5,pg.no.20
2 x=[1 2 4 8 16 32 64];
3 y=[0 0 0 1 0 0 0];
4 z=x.*y;
5 a=0;
6 for i=1:length(z)
7     a=a+z(i);
8 end
9 z,a//a=summation of z
```

#### Scilab code Exa 2.6 Discrete time Harmonics and Periodicity

```
1 //example 2.6 pg.no.23
2 function[p]=period(x)
3 for i=2:length(x)
4 v=i
```

```
if (abs(x(i)-x(1))<0.00001)
5
6
             k=2
             for j=i+1:i+i
7
                  if (abs(x(j)-x(k))<0.00001)
8
9
                       v = v + 1
10
                  end
11
             k=k+1;
12
             end
13
         end
14
        if (v==(2*i)) then
15
             break
16
        end
17 end
18 p = i - 1
19 endfunction
20 for i=1:60
21
        x1(i) = \cos((2*\%pi*8*i)/25);
22 \text{ end}
23 \text{ for } i=1:60
        x2(i) = exp(\%i*0.2*i*\%pi) + exp(-\%i*0.3*i*\%pi);
24
25 end
26 \text{ for } i=1:45
        x3(i)=2*cos((40*\%pi*i)/75)+sin((60*\%pi*i)/75);
27
28 end
29 \text{ period}(x1)
30 \text{ period}(x2)
31 period(x3)
```

check Appendix AP 1 for dependency:

Aliasfrequency.sci

Scilab code Exa 2.7 Aliasing and its effects

```
1 //example 2.7.pg.no.27
2 f=100;
```

```
3 s = 240;
4 s1=s;
5 aliasfrequency(f,s)
6 \text{ s} = 140;
7 s1=s;
8 aliasfrequency(f,s,s1)
9 s = 90;
10 \text{ s1=s};
11 aliasfrequency(f,s,s1)
12 \text{ s} = 35;
13 \text{ s1=s};
14 aliasfrequency(f,s,s1)
      check Appendix AP 1 for dependency:
      Aliasfrequency.sci
   Scilab code Exa 2.8 Signal Reconstruction
1 f=100;
2 s = 210;
3 s1=420;
4 aliasfrequency(f,s,s1)
5 s = 140;
6 aliasfrequency(f,s,s1)
```

### Chapter 3

### Response of Digital Filters

Scilab code Exa 3.5 FIR filter response

```
1 //Response of non-recursive Filters
2 for i=1:4
3     x(i)=0.5^i;
4 end
5 x1=[0;1;x(1:2)]
6 for i=1:4
7     y(i)=2*x(i)-3*x1(i);
8 end
9 y(1),y(2)
```

Scilab code Exa 3.19a Analytical Evaluation of Discrete Convolution

```
1 // Analytical evaluation of Discrete Convolution
2 clear; close; clc;
3 max_limit=10;
4 h=ones(1,max_limit);
5 n2=0:length(h)-1;
6 x=h;
```

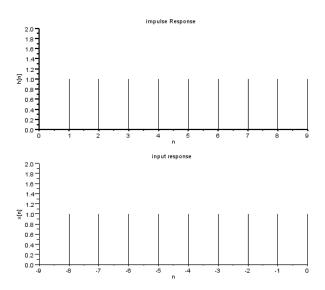


Figure 3.1: Analytical Evaluation of Discrete Convolution

```
7 n1 = -length(x) + 1:0;
8 y = convol(x,h);
9 n=-length(x)+1:length(h)-1;
10 a=gca();
11 subplot(211);
12 plot2d3('gnn',n2,h)
13 xtitle('impulse Response', 'n', 'h[n]');
14 a.thickness=2;
15 a.y_location="origin";
16 subplot (212);
17 plot2d3('gnn',n1,x)
18 a.y_location="origin";
19 xtitle('input response', 'n', 'x[n]');
20 xset("window",1);
21 a=gca();
22 plot2d3('gnn',n,y)
23 xtitle('output response', 'n', 'y[n]');
```

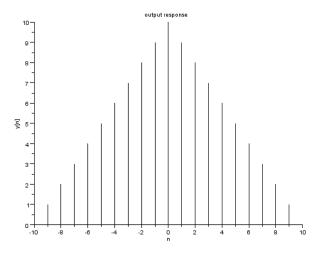


Figure 3.2: Analytical Evaluation of Discrete Convolution

#### Scilab code Exa 3.19b Analytical Evaluation of Discrete Convolution

```
1 clear; close; clc;
2 max_limit=10;
3 for n=1:max_limit
4     h(n)=(0.4)^n;
5 end
6 n2=0:length(h)-1;
7 for n=1:max_limit
8     x(n)=(0.8)^n;
9 end
10 n1=-length(x)+1:0;
11 y=convol(x,h)
12 n=-length(x)+1:length(h)-1;
13 a=gca();
```

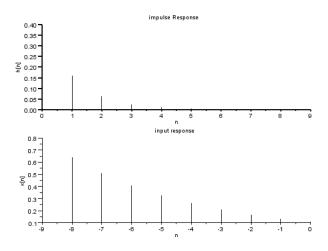


Figure 3.3: Analytical Evaluation of Discrete Convolution

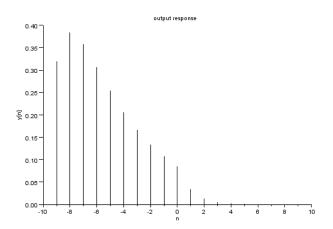


Figure 3.4: Analytical Evaluation of Discrete Convolution

Scilab code Exa 3.19c Analytical Evaluation of Discrete Convolution

```
1 // Analytical Evaluation of Discrete convolution
2 clear; close; clc;
3 max_limit=5;
4 h(1)=0;
5 for n=2:max_limit
6
       h(n) = 0.8^n;
7 end
8 n2=0:length(h)-1;
9 x = [0 \text{ ones} (1: max\_limit)]
10 n1 = -length(x) + 1:0;
11 y = convol(x,h);
12 n=-length(x)+1:length(h)-1;
13 a=gca();
14 subplot (211);
15 plot2d3('gnn',n2,h)
16 xtitle('impulse Response', 'n', 'h[n]');
17 a.thickness=2;
18 a.y_location="origin";
```

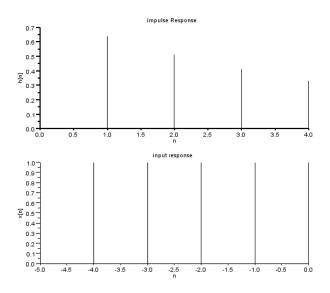


Figure 3.5: Analytical Evaluation of Discrete Convolution

```
19    a=gca();
20    subplot(212);
21    plot2d3('gnn',n1,x)
22    a.y_location="origin";
23    xtitle('input response','n','x[n]');
24    xset("window",1);
25    a=gca();
26    plot2d3('gnn',n,y)
27    a.y_location="origin";
28    a.x_location="origin";
29    xtitle('output response','n','y[n]');
```

Scilab code Exa 3.20a Properties of Convolution

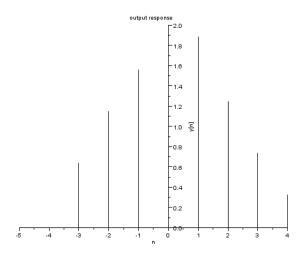


Figure 3.6: Analytical Evaluation of Discrete Convolution

```
1 // properties of convolution
2 x=[1 2 3 4 5];
3 h=[1 zeros(1:5)];
4 a=convol(x,h);
5 b=convol(h,x);
6 a==b
```

#### Scilab code Exa 3.20b Properties of Convolution

```
1 //Convolution with Step Function
2 x=[1 2 3 4 5];
3 h=[ones(1:5)];
4 a=convol(h,x);
5 b(1)=a(1);
6 for i=2:length(x)
7 b(i)=b(i-1)+x(i);
8 end
```

```
9 disp(a(1:length(x)),b,'Step Response is running sum of impulses can be seen below');
```

#### Scilab code Exa 3.21a Convolution of finite length Signals

```
1 //convolution of finite length signals
2 clear; close; clc;
3 max_limit=10;
4 h = [1 2 2 3];
5 n2=0:length(h)-1;
6 x = [2 -1 3];
7 \text{ n1=0:} length(x)-1;
8 y = convol(x,h);
9 \text{ n=0:length(h)+length(x)-2;}
10 a=gca();
11 subplot (211);
12 plot2d3('gnn',n2,h)
13 xtitle('impulse Response', 'n', 'h[n]');
14 a.thickness=2;
15 a.y_location="origin";
16 a=gca();
17 subplot (212);
18 plot2d3('gnn',n1,x)
19 a.y_location="origin";
20 a.x_location="origin";
21 xtitle('input response', 'n', 'x[n]');
22 xset("window",2);
23 a=gca();
24 plot2d3('gnn',n,y)
25 a.y_location="origin";
26 xtitle('output response', 'n', 'y[n]');
```

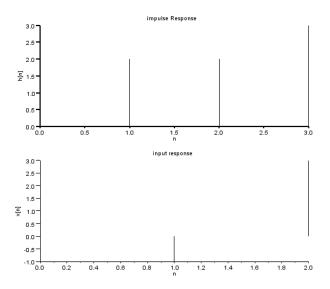


Figure 3.7: Convolution of finite length Signals

#### Scilab code Exa 3.21b Convolution of finite length Signals

```
1 clear; close; clc;
2 max_limit=10;
3 h=[2 5 0 4];
4 n2=-2:length(h)-3;
5 x=[4 1 3];
6 n1=-1:length(x)-2;
7 y=convol(x,h);
8 n=-3:length(x)+length(h)-5;
9 a=gca();
10 subplot(211);
11 plot2d3('gnn',n2,h)
12 xtitle('impulse Response', 'n', 'h[n]');
13 a.thickness=2;
```

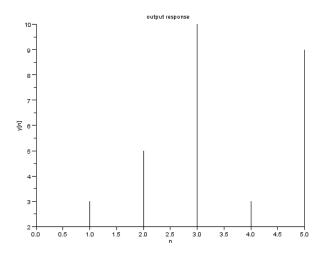


Figure 3.8: Convolution of finite length Signals

```
14 a.y_location="origin";
15 a=gca();
16 subplot(212);
17 plot2d3('gnn',n1,x)
18 a.y_location="origin";
19 xtitle('input response', 'n', 'x[n]');
20 xset("window",1);
21 a=gca();
22 plot2d3('gnn',n,y)
23 a.y_location="origin";
24 a.x_location="origin";
25 xtitle('output response', 'n', 'y[n]');
```

 ${\bf Scilab\ code\ Exa\ 3.21c\ } {\bf Convolution\ of\ finite\ length\ Signals}$ 

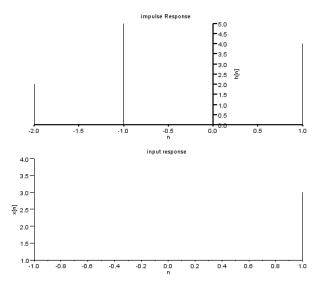


Figure 3.9: Convolution of finite length Signals

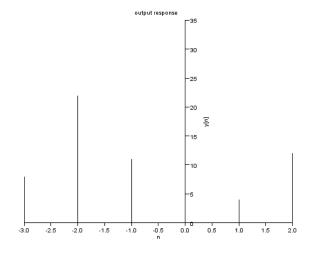


Figure 3.10: Convolution of finite length Signals

```
1 clear; close; clc;
2 max_limit=10;
3 h = [1/2 1/2 1/2];
4 n2=0:length(h)-1;
5 x = [2 4 6 8 10];
6 \text{ n1=0:length}(x)-1;
7 y = convol(x,h);
8 n=0:length(x)+length(h)-2;
9 a=gca();
10 subplot (211);
11 plot2d3('gnn',n2,h);
12 xtitle('impulse Response', 'n', 'h[n]');
13 a.thickness=2;
14 a.y_location="origin";
15 a=gca();
16 subplot (212);
17 plot2d3('gnn',n1,x);
18 a.y_location="origin";
19 xtitle('input response', 'n', 'x[n]');
20 xset("window",1);
21 a=gca();
22 plot2d3('gnn',n,y)
23 a.y_location="origin";
24 a.x_location="origin";
25 xtitle('output response', 'n', 'y[n]');
```

#### Scilab code Exa 3.22 Convolution of finite length Signals

```
1 max_limit=10;
2 h=[2 5 0 4];
3 n2=0:length(h)-1;
```

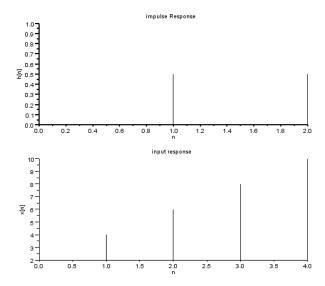


Figure 3.11: Convolution of finite length Signals

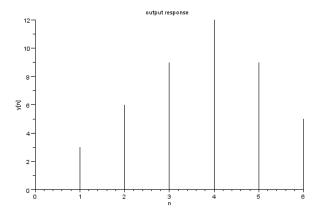


Figure 3.12: Convolution of finite length Signals

```
4 x = [4 1 3];
5 \text{ n1=0:length}(x)-1;
6 y = convol(x,h);
7 \text{ n=0:length(x)+length(h)-2;}
8 a = gca();
9 subplot(211);
10 plot2d3('gnn',n2,h)
11 xtitle('impulse Response', 'n', 'h[n]');
12 a.thickness=2;
13 a.y_location="origin";
14 a=gca();
15 subplot (212);
16 plot2d3('gnn',n1,x)
17 a.y_location="origin";
18 a.x_location="origin";
19 xtitle('input response', 'n', 'x[n]');
20 xset ("window", 1);
21 a=gca();
22 plot2d3('gnn',n,y)
23 a.y_location="origin";
24 a.x_location="origin";
25 xtitle('output response', 'n', 'y[n]');
```

Scilab code Exa 3.23 effect of Zero Insertion, Zero Padding on convol.

```
1 //convolution by polynomial method
2 x=[4 1 3];
3 h=[2 5 0 4];
4 z=%z;
5 n=length(x)-1:-1:0;
6 X=x*z^n';
```

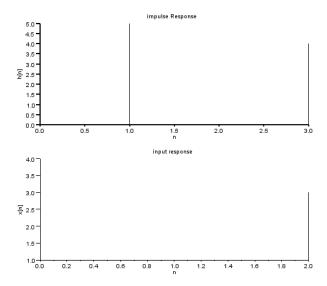


Figure 3.13: Convolution of finite length Signals

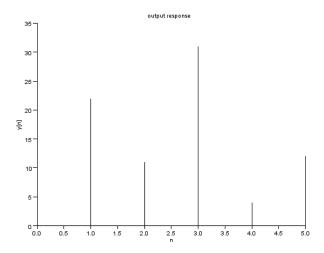


Figure 3.14: Convolution of finite length Signals

```
7 n1=length(h)-1:-1:0;
8 H=h*z^n1';
9 y=X*H
10 //effect of zero insertion on convolution
11 h=[2 0 5 0 0 0 4];
12 x=[4 0 1 0 3];
13 y=convol(x,h)
14 //effect of zero padding on convolution
15 h=[2 5 0 4 0 0];
16 x=[4 1 3 0];
17 y=convol(x,h)
```

## Scilab code Exa 3.25 Stability and Causality

```
1 //concepts based on stability and Causality
2 function[]=stability(X)
3 if (abs(roots(X))<1)</pre>
       disp("given system is stable")
4
5 else
       disp("given system is not stable")
6
7 end
8 endfunction
9 x = [1 -1/6 -1/6];
10 z = \%z;
11 n = length(x) - 1: -1:0;
12 //characteristic eqn is
13 X = x * (z)^n
14 stability(X)
15 x = [1 -1];
16 \text{ n=length}(x)-1:-1:0;
17 //characteristic eqn is
18 X = x * (z)^n
19 stability(X)
20 x = [1 -2 1];
21 n = length(x) - 1: -1:0;
```

```
22 //characteristic eqn is
23 X=x*(z)^n'
24 stability(X)
```

## Scilab code Exa 3.26 Response to Periodic Inputs

```
1 //Response of periodic inputs
2 function[p]=period(x)
3 for i=2:length(x)
4
       v = i
        if (abs(x(i)-x(1))<0.00001)
5
6
            k=2
7
            for j=i+1:i+i
8
                 if (abs(x(j)-x(k))<0.00001)
9
                     v = v + 1
10
                 end
11
            k=k+1;
12
            end
13
         end
14
        if (v==(2*i)) then
15
            break
16
        end
17 \text{ end}
18 p = i - 1
19 endfunction
20 x = [1 2 -3 1 2 -3 1 2 -3];
21 h=[1 1];
22 y = convol(x,h)
23 y(1) = y(4);
24 period(x)
25 period(y)
26 h=[1 1 1];
27 y = convol(x,h)
```

#### Scilab code Exa 3.27 Periodic Extension

```
1 //to find periodic extension
2 x=[1 5 2;0 4 3;6 7 0];
3 y=[0 0 0];
4 for i=1:3
5     for j=1:3
6         y(i)=y(i)+x(j,i);
7     end
8 end
9 y
```

### Scilab code Exa 3.28 System Response to Periodic Inputs

```
1 //method of wrapping to fing convolution of periodic
       signal with one period
2 x = [1 2 -3];
3 h = [1 1];
4 y1 = convol(h,x)
5 y1 = [y1, zeros(5:9)]
6 y2=[y1(1:3);y1(4:6);y1(7:9)];
7 y = [0 \ 0 \ 0];
8 for i=1:3
       for j=1:3
9
            y(i)=y(i)+y2(j,i);
10
11
       end
12 end
13 y
14 x = [2 1 3];
15 h=[2 1 1 3 1];
16 y1 = convol(h,x)
17 y1 = [y1, zeros(8:9)]
```

```
18 y2=[y1(1:3);y1(4:6);y1(7:9)];

19 y=[0 0 0];

20 for i=1:3

21 for j=1:3

22 y(i)=y(i)+y2(j,i);

23 end

24 end

25 y
```

#### Scilab code Exa 3.29 Periodic Convolution

```
1 // periodic or circular convolution
2 x = [1 \ 0 \ 1 \ 1];
3 h = [1 2 3 1];
4 y1 = convol(h,x)
5 y1 = [y1, zeros(8:12)];
6 y2=[y1(1:4);y1(5:8);y1(9:12)];
7 y = [0 \ 0 \ 0 \ 0];
8 \text{ for } i=1:4
9
        for j=1:3
             y(i)=y(i)+y2(j,i);
10
11
        end
12 end
13 y
```

#### Scilab code Exa 3.30 Periodic Convolution by Circulant Matrix

```
// periodic convolution by circulant matrix
x=[1 0 2];
h=[1;2;3];
// generation of circulant matrix
c(1,:)=x;
for i=2:length(x)
```

```
7      c(i,:)=[x(length(x):length(x)-i),x(1:length(x)-i
          )]
8    end
9    c'
```

### Scilab code Exa 3.32 Deconvolution By polynomial Division

```
1 //deconvolution by polynomial division
2 x=[2 5 0 4];
3 y=[8 22 11 31 4 12];
4 z=%z
5 n=length(x)-1:-1:0;
6 X=x*(z)^n'
7 n1=length(y)-1:-1:0;
8 Y=y*(z)^n1'
9 h=Y/X
```

#### Scilab code Exa 3.33 Autocorrelation and Cross Correlation

```
//discrete auto correlation and cross correlation
x=[2 5 0 4];
h=[3 1 4];
x1=x(length(x):-1:1)
h1=h(length(h):-1:1)
rxhn=convol(x,h1)
rhxn=convol(x1,h)
rhxn=rhxn(length(rhxn):-1:1)
//we observe that rhxn1=rxhn
x=[3 1 -4];
x1=x(length(x):-1:1)
rxxn=convol(x,x1)
//we observe that rxxn is even symmetric about origin
```

#### Scilab code Exa 3.35 Periodic Autocorrelation and Cross Correlation

```
1 //discrete periodic auto correlation and cross
      correlation
2 x = [2 5 0 4];
3 h = [3 1 -1 2];
4 x1=x(length(x):-1:1);
5 h1=h(length(h):-1:1);
6 rxhn=convol(x,h1)
7 rhxn=convol(x1,h)
8 rxxn=convol(x,x1)
9 rhhn=convol(h,h1)
10 y1=[rxhn, zeros(8:12)];
11 y2=[y1(1:4);y1(5:8);y1(9:12)];
12 y3=[rhxn, zeros(8:12)];
13 y4 = [y3(1:4); y3(5:8); y3(9:12)];
14 y5=[rxxn, zeros(8:12)];
15 y6 = [y5(1:4); y5(5:8); y5(9:12)];
16 y7=[rhhn, zeros(8:12)];
17 y8 = [y7(1:4); y7(5:8); y7(9:12)];
18 \text{ rxhp} = [0 \ 0 \ 0 \ 0];
19 rhxp=[0 0 0 0];
20 \text{ rxxn} = [0 \ 0 \ 0 \ 0];
21 rhhp=[0 0 0 0];
22 \text{ for } i=1:4
23
         for j=1:3
            rhxp(i)=rhxp(i)+y4(j,i);
24
            rxhp(i)=rxhp(i)+y2(j,i);
25
            rxxn(i) = rxxn(i) + y6(j,i);
26
27
            rhhp(i)=rhhp(i)+y8(j,i);
28
       end
29 end
30 rxhp
31 rhxp
```

32 rxxn

33 rhhp

# Chapter 4

# z Transform Analysis

Scilab code Exa 4.1b z transform of finite length sequences

```
1 function[za] = ztransfer(sequence,n)
2         z = poly(0, 'z', 'r')
3         za = sequence * (1/z) ^n'
4 endfunction
5 x1 = [2 1 -5 4];
6 n = -1: length(x1) - 2;
7 ztransfer(x1,n)
```

#### Scilab code Exa 4.4a Pole Zero Plots

```
1 //Pole-Zero plots
2 z=%z;
3 az=2*z*(z+1);
4 bz=(z-1/3)*((z^2)+1/4)*((z^2)+4*z+5);
5 poles=roots(bz)
6 zeroes=roots(az)
7 h=az/bz
```

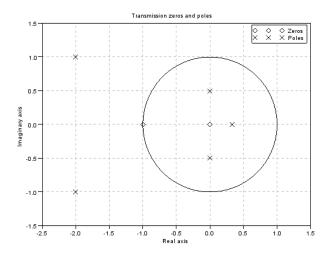


Figure 4.1: Pole Zero Plots

```
8 plzr(h)
```

# Scilab code Exa 4.4b Pole Zero plots

```
1 //Pole-Zero plots
2 z=%z;
3 az=z^4+4.25*z^2+1;
4 bz=z^4;
5 poles=roots(bz)
6 zeroes=roots(az)
7 h=az/bz
8 plzr(h)
```

Scilab code Exa 4.8 Stability of Recursive Filters

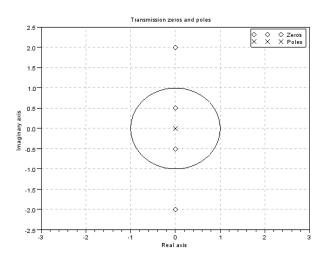


Figure 4.2: Pole Zero plots

```
1 //stability of recursive filter
\frac{1}{2} / | for roc : /z/ > /a/
3 a=input('enter the value of alpha')
4 z = \%z;
5 \text{ H=z/(z-a)};
6 if (abs(a)<1)
        disp("system is stable")
8
  else
       disp("system is not stable")
9
10
    end
    //for roc:/z/</a/
11
    if (abs(a)>1)
12
       disp("system is stable")
13
14 else
        disp("system is not stable")
15
16
    end
```

Scilab code Exa 4.9 Inverse Systems

```
1 //inverse systems
2 z=%z;
3 H=(1+2*(z^(-1)))/(1+3*(z^(-1)));
4 //inverse of H is
5 H1=1/H
6 H=1+2*(z^(-1))+3*(z^(-2));
7 H1=1/H
```

### Scilab code Exa 4.10 Inverse Transform of sequences

```
//inverse transform of sequences
//(a)X(z)=3z^-1+5z^-3+2z^-4
z=%z;
X1=[3*z^-1;0;5*z^-3;2*z^-4];
n1=1:4;
ZI=z^n1';
x1=numer(X1.*ZI);
disp(x1,"x[n]=");
//(b)X(z)=2z^2-5z+5z^-1-2z^-2
X2=[2*z^2;-5*z;0;5*z^-1;-2*z^-2];
n2=-2:2;
ZI=z^n2';
x2=numer(X2.*ZI);
disp(x2,"x[n]=");
```

#### Scilab code Exa 4.11 Inverse Transform by Long Division

```
1 //inverse transform by long division
2 z=%z;
3 x=ldiv(z-4,1-z+z^2,5)
```

#### Scilab code Exa 4.12 Inverse transform of Right sided sequences

```
1 //inverse z transforms from standard transforms 2 z=%z; 3 xz=z/((z-0.5)*(z-0.25)); 4 yz=xz/z; 5 pfss(yz) 6 //hence x[n]=-4(0.25)^n*un+4(0.5)^n*un; 7 xz=1/((z-0.5)*(z-0.25)); 8 yz=xz/z; 9 pfss(yz) 10 //hence x[n]=-4(0.25)^n-1*u[n-1]+4(0.5)^n-1*u[n-1];
```

### Scilab code Exa 4.20 z Transform of Switched periodic Signals

```
1 //z transform of switched periodic signals
2 z = \%z;
3 / sol. for 4.20a
4 x1 = [0 1 2];
5 n=0:2;
6 N = 3;
7 x1z=x1*(1/z)^n
8 xz = x1z/(1-z^-N)
9 // sol. for 4.20b
10 x1 = [0 \ 1 \ 0 \ -1];
11 n=0:3;
12 N = 4;
13 x1z=x1*(1/z)^n
14 xz = x1z/(1-z^-N)
15 / sol. for 4.20c
16 \text{ xz} = (2+z^-1)/(1-z^-3);
17 	 x1z = numer(xz)
18 //thus first period of xn is [2 1 0]
19 / sol. for 4.20d
20 xz = (z^-1-z^-4)/(1-z^-6);
```

```
21 x1z=numer(xz)
22 //thus first period of xn is [0 1 0 0 -1 0]
```

# Chapter 5

# Frequency Domain Analysis

Scilab code Exa 5.1c DTFT from Defining Relation

```
1 //DTFT of x[n]=(a)^n*u[n]
2 clear;
3 clc; close;
4 //DTS signal
5 a1=0.5;
6 \quad a2 = -0.5;
7 max_limit=10;
8 for n=0:max_limit-1
     x1(n+1) = (a1^n);
10
     x2(n+1)=(a2^n);
11 end
12 n=0:max_limit-1;
13 //discrete time fourier transform
14 wmax=2*\%pi;
15 K=4;
16 \text{ k=0:}(K/1000):K;
17 W=k*wmax/K;
18 x1=x1,
19 x2=x2,
20 XW1 = x1 * exp(%i*n'*W);
21 XW2=x2*exp(%i*n'*W);
```

```
22 XW1_Mag = abs(XW1);
23 XW2\_Mag=abs(XW2);
24 W=[-mtlb_fliplr(W), W(2:1001)]; //omega form
25 XW1_Mag=[mtlb_fliplr(XW1_Mag),XW1_Mag(2:1001)];
26 XW2_Mag=[mtlb_fliplr(XW2_Mag),XW2_Mag(2:1001)];
27 [XW1_phase,db]=phasemag(XW1);
28 [XW2_phase,db]=phasemag(XW2);
29 XW1_phase=[-mtlb_fliplr(XW1_phase),XW1_phase(2:1001)
30 XW2_phase=[-mtlb_fliplr(XW2_phase),XW2_phase(2:1001)
      ];
31
32 //plot for a>0
33 figure
34 subplot (3,1,1);
35 plot2d3('gnn',n,x1)
36 xtitle('Discrete time sequencex[n] a>0')
37 subplot (3,1,2);
38 \ a = gca();
39 a.y_location="origin";
40 a.x_location="origin";
41 plot2d3(W,XW1_Mag);
42 title('magnitude Response abs(exp(jw))')
43 subplot(3,1,3);
44 \ a = gca();
45 a.y_location="origin";
46 a.x_location="origin";
47 plot2d(W,XW1_phase);
48 title('magnitude Response abs(exp(jw))')
49 //plot for a<0
50 figure
51 subplot (3,1,1);
52 plot2d3('gnn',n,x2);
53 xtitle('Discrete Time sequence x[n] for a>0')
54 subplot (3,1,2);
55 a = gca();
56 a.y_location="origin";
57 a.x_location="origin";
```

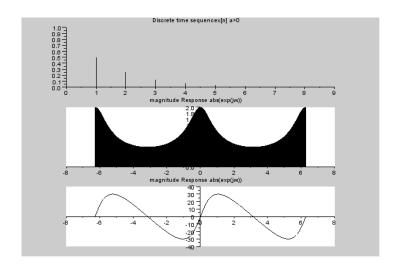


Figure 5.1: DTFT from Defining Relation

```
58 plot2d(W,XW2_Mag);
59 title('Magnitude Response abs(X(jw))')
60 subplot(3,1,3);
61 a=gca();
62 a.y_location="origin";
63 a.x_location="origin";
64 plot2d(W,XW2_phase);
65 title('phase Response<(X(jw))')</pre>
```

Scilab code Exa 5.3a Some DTFT pairs using properties

```
1 //DTFT of x[n]=n*(a)^n*u[n]
2 clear;
3 clc; close;
```

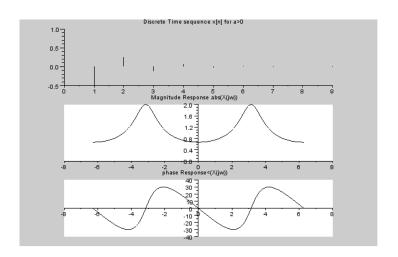


Figure 5.2: DTFT from Defining Relation

```
4 //DTS signal
5 a1=0.5;
6 \quad a2 = -0.5;
7 max_limit=10;
8 for n=0:max_limit-1
     x1(n+1)=n*(a1^n);
10
     x2(n+1)=n*(a2^n);
11 end
12 \quad n=0:max_limit-1;
13 //discrete time fourier transform
14 wmax=2*\%pi;
15 K=4;
16 k=0:(K/1000):K;
17 W=k*wmax/K;
18 \times 1 = \times 1;
19 x2=x2;
20 XW1 = x1 * exp(%i*n'*W);
21 XW2 = x2 * exp(%i*n'*W);
22 \text{ XW1\_Mag=abs}(XW1);
23 XW2\_Mag=abs(XW2);
```

```
24 W=[-mtlb_fliplr(W), W(2:1001)]; //omega form
25 XW1_Mag=[mtlb_fliplr(XW1_Mag),XW1_Mag(2:1001)];
26 XW2_Mag=[mtlb_fliplr(XW2_Mag),XW2_Mag(2:1001)];
27 [XW1_phase,db]=phasemag(XW1);
28 [XW2_phase,db]=phasemag(XW2);
29 XW1_phase=[-mtlb_fliplr(XW1_phase),XW1_phase(2:1001)
     ];
30 XW2_phase=[-mtlb_fliplr(XW2_phase),XW2_phase(2:1001)
     ];
31
32 //plot for a>0
33 figure
34 subplot (3,1,1);
35 plot2d3('gnn',n,x1)
36 xtitle ('Discrete time sequencex[n] a>0')
37 subplot(3,1,2);
38 \quad a = gca();
39 a.y_location="origin";
40 a.x_location="origin";
41 plot2d3(W, XW1_Mag);
42 title ('magnitude Response abs (exp(jw))')
43 subplot (3,1,3);
44 \ a = gca();
45 a.y_location="origin";
46 a.x_location="origin";
47 plot2d(W,XW1_phase);
48 title('magnitude Response abs(exp(jw))')
49 //plot for a<0
50 figure
51 subplot (3,1,1);
52 plot2d3('gnn',n,x2);
53 xtitle('Discrete Time sequence x[n] for a>0')
54 subplot (3,1,2);
55 \quad a = gca();
56 a.y_location="origin";
57 a.x_location="origin";
58 plot2d(W,XW2_Mag);
59 title ('Magnitude Response abs(X(jw))')
```

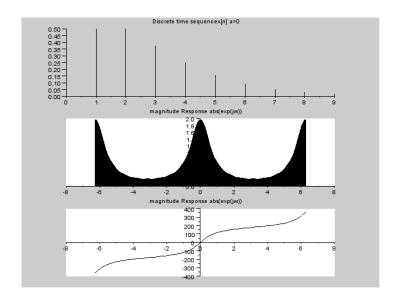


Figure 5.3: Some DTFT pairs using properties

```
60 subplot(3,1,3);
61 a=gca();
62 a.y_location="origin";
63 a.x_location="origin";
64 plot2d(W,XW2_phase);
65 title('phase Response <(X(jw))')
```

# Scilab code Exa 5.3b Some DTFT pairs using properties

```
1 //DTFT of x[n]=n*(a)^n*u[n]
2 clear;
3 clc; close;
4 //DTS signal
5 a1=0.5;
```

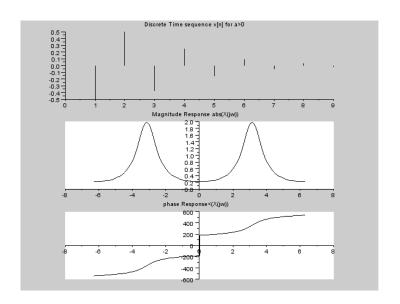


Figure 5.4: Some DTFT pairs using properties

```
6 \quad a2 = -0.5;
7 max_limit=10;
8 for n=0:max_limit-1
     x1(n+1) = (n+1)*(a1^n);
     x2(n+1)=(n+1)*(a2^n);
10
11 end
12 n=0:max_limit-1;
13 //discrete time fourier transform
14 wmax=2*%pi;
15 K=4;
16 \text{ k=0:}(K/1000):K;
17 W=k*wmax/K;
18 \times 1 = \times 1;
19 x2=x2;
20 XW1 = x1 * exp(%i*n'*W);
21 XW2 = x2 * exp(%i*n'*W);
22 XW1_Mag = abs(XW1);
23 XW2\_Mag=abs(XW2);
24 W=[-mtlb_fliplr(W),W(2:1001)];//omega form
25 XW1_Mag=[mtlb_fliplr(XW1_Mag),XW1_Mag(2:1001)];
```

```
26 XW2_Mag=[mtlb_fliplr(XW2_Mag),XW2_Mag(2:1001)];
27 [XW1_phase,db]=phasemag(XW1);
28 [XW2_phase,db]=phasemag(XW2);
29 XW1_phase=[-mtlb_fliplr(XW1_phase),XW1_phase(2:1001)
      ];
30 XW2_phase=[-mtlb_fliplr(XW2_phase),XW2_phase(2:1001)
      ];
31
32 //plot for a>0
33 figure
34 subplot (3,1,1);
35 plot2d3('gnn',n,x1)
36 xtitle('Discrete time sequencex[n] a>0')
37 subplot (3,1,2);
38 \ a = gca();
39 a.y_location="origin";
40 a.x_location="origin";
41 plot2d3(W,XW1_Mag);
42 title('magnitude Response abs(exp(jw))')
43 subplot (3,1,3);
44 a=gca();
45 a.y_location="origin";
46 a.x_location="origin";
47 plot2d(W,XW1_phase);
48 title('magnitude Response abs(exp(jw))')
49 //plot for a<0
50 figure
51 subplot(3,1,1);
52 plot2d3('gnn',n,x2);
53 xtitle('Discrete Time sequence x[n] for a>0')
54 subplot (3,1,2);
55 \quad a = gca();
56 a.y_location="origin";
57 a.x_location="origin";
58 plot2d(W,XW2_Mag);
59 title ('Magnitude Response abs(X(jw))')
60 subplot(3,1,3);
61 a=gca();
```

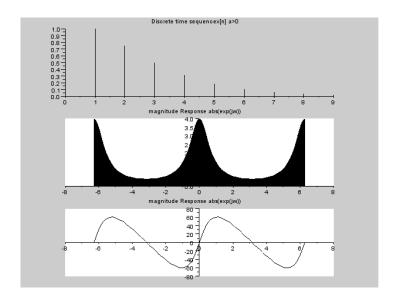


Figure 5.5: Some DTFT pairs using properties

```
62 a.y_location="origin";
63 a.x_location="origin";
64 plot2d(W,XW2_phase);
65 title('phase Response<(X(jw))')</pre>
```

# Scilab code Exa 5.3d Some DTFT pairs using properties

```
1  //DTFT of x[n]=a^abs(n)
2  a=0.5;
3  n=-9:9;
4  x=a^abs(n);
5  // Discrete time Fourier Transform
```

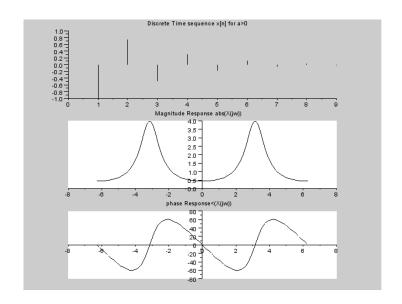


Figure 5.6: Some DTFT pairs using properties

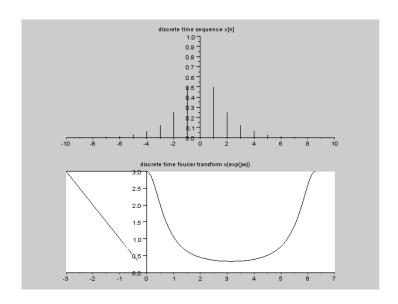


Figure 5.7: Some DTFT pairs using properties

```
6 \text{ k=0:} (4/1000):4;
7 w = (2*\%pi*k)/4;
8 xw = x * exp(%i*n'*w);
9 xw_mag=real(xw);
10 w=[-mtlb_fliplr(xw_mag),w(2:1001)];
11 xw_mag=[mtlb_fliplr(xw_mag),xw_mag(2:1001)];
12 figure
13 subplot(2,1,1);
14 a=gca();
15 a.x_location="origin";
16 a.y_location="origin";
17 plot2d3('gnn',n,x);
18 xtitle('discrete time sequence x[n]');
19 subplot(2,1,2);
20 a=gca();
21 a.x_location="origin";
22 a.y_location="origin";
23 plot2d(w,xw_mag);
24 title('discrete time fourier transform x(exp(jw))');
```

#### Scilab code Exa 5.3e Some DTFT pairs using properties

```
1 //DTFT of x[n]=n*(a)^n*u[n]
2 clear;
3 clc; close;
4 //DTS signal
5 a1=0.5;
6 a2=-0.5;
7 max_limit=10;
8 for n=0:max_limit-1
9 x1(n+1)=4*(a1^(n+3));
10 x2(n+1)=4*(a2^(n+3));
11 end
12 n=0:max_limit-1;
13 //discrete time fourier transform
```

```
14 wmax=2*\%pi;
15 K=4;
16 k=0:(K/1000):K;
17 W=k*wmax/K;
18 \times 1 = \times 1;
19 x2=x2;
20 XW1 = x1 * exp(%i*n'*W);
21 XW2=x2*exp(%i*n'*W);
22 \times XW1_Mag = abs(XW1);
23 XW2\_Mag=abs(XW2);
24 W=[-mtlb_fliplr(W),W(2:1001)];//omega form
25 XW1_Mag=[mtlb_fliplr(XW1_Mag),XW1_Mag(2:1001)];
26 XW2_Mag=[mtlb_fliplr(XW2_Mag), XW2_Mag(2:1001)];
27 [XW1_phase,db]=phasemag(XW1);
28 [XW2_phase,db]=phasemag(XW2);
29 XW1_phase=[-mtlb_fliplr(XW1_phase),XW1_phase(2:1001)
      ];
30 XW2_phase=[-mtlb_fliplr(XW2_phase),XW2_phase(2:1001)
      ];
31
32 //plot for a>0
33 figure
34 subplot (3,1,1);
35 plot2d3 ('gnn',n,x1)
36 xtitle('Discrete time sequencex[n] a>0')
37 subplot (3,1,2);
38 \ a = gca();
39 a.y_location="origin";
40 a.x_location="origin";
41 plot2d3(W, XW1_Mag);
42 title('magnitude Response abs(exp(jw))')
43 subplot (3,1,3);
44 \ a = gca();
45 a.y_location="origin";
46 a.x_location="origin";
47 plot2d(W,XW1_phase);
48 title('magnitude Response abs(exp(jw))')
49 //plot for a<0
```

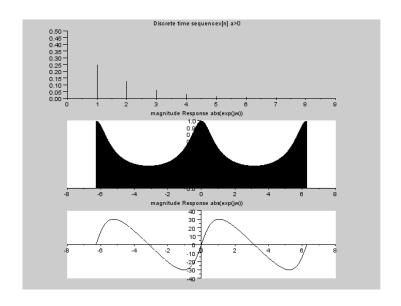


Figure 5.8: Some DTFT pairs using properties

```
50 figure
51 subplot(3,1,1);
52 plot2d3('gnn',n,x2);
53 xtitle('Discrete Time sequence x[n] for a>0')
54 subplot(3,1,2);
55 \quad a=gca();
56 a.y_location="origin";
57 a.x_location="origin";
58 plot2d(W,XW2_Mag);
59 title('Magnitude Response abs(X(jw))')
60 subplot(3,1,3);
61 \quad a = gca();
62 a.y_location="origin";
63 a.x_location="origin";
64 plot2d(W,XW2_phase);
65 title('phase Response <(X(jw))')
```

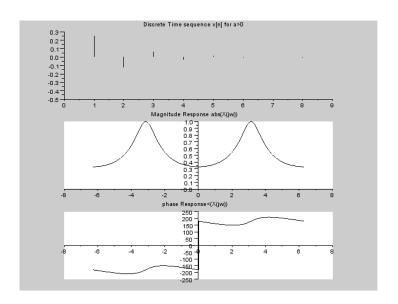


Figure 5.9: Some DTFT pairs using properties

# Scilab code Exa 5.4 DTFT of periodic Signals

```
1 //DTfT of periodic signals
2 x=[3 2 1 2];//one period of signal
3 n=0:3;
4 k=0:3;
5 x1=x*exp(%i*n'*2*k*%pi/4)
6 dtftx=abs(x1)
7 x=[3 2 1 2 3 2 1 2 3];
8 n=-4:4;
9 a=gca();
10 a.y_location="origin";
11 a.x_location="origin";
12 plot2d3('gnn',n,x);
13 xtitle('discrete periodic time signal');
```

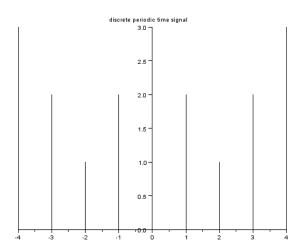


Figure 5.10: DTFT of periodic Signals

```
14 x2=[dtftx dtftx 8];
15 a=gca();
16 xset('window',1);
17 a.x_location="origin";
18 a.y_location="origin";
19 plot2d3('gnn',n,x2);
20 xtitle('DTFT of discrete periodic signal');
```

## Scilab code Exa 5.5 The DFT, DFS and DTFT

```
1 x=[1 0 2 0 3];//one period of signal
2 n=0:4;
3 k=0:4;
4 x1=x*exp(%i*n'*2*k*%pi/4)
```

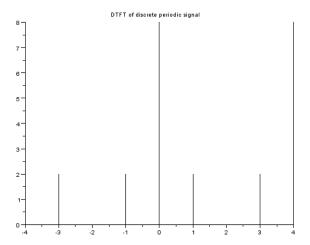


Figure 5.11: DTFT of periodic Signals

```
5 DTFTx=abs(x1)
6 DFT=dft(x,-1)
7 DFS=DFT/5
```

# Scilab code Exa 5.7 Frequency Response of Recursive Filter

```
1 a=0.5;b=1;
2 n=0:50;
3 h=b*(a^n);
4 // Discrete-Time Fourier transform
5 K=500;
6 k=-250:1:250;
7 w=%pi*k/K;
8 H=h*exp(-%i*n'*w);
9 // caluculation of phase and magnitude of h(z)
10 [phase_H,m]=phasemag(H);
```

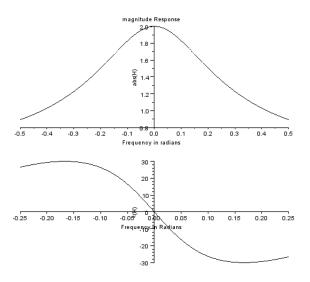


Figure 5.12: Frequency Response of Recursive Filter

```
11 H=abs(H);
12 a=gca();
13 subplot(2,1,1);
14 a.y_location="origin";
15 plot2d(w/%pi,H);
16 xlabel('Frequency in radians')
17 ylabel('abs(H)')
18 title('magnitude Response')
19 subplot(2,1,2);
20 a=gca();
21 a.x_location="origin";
22 a.y_location="origin";
23 plot2d(w/(2*%pi),phase_H)
24 xlabel('Frequency in Radians');
25 ylabel('<(H)')
26 title('Phase Response'))
```

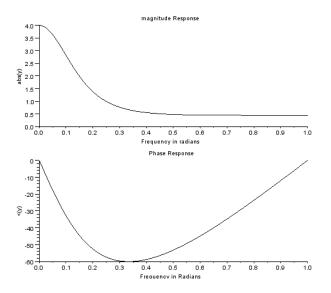


Figure 5.13: The DTFT in System Analysis

## Scilab code Exa 5.8a The DTFT in System Analysis

```
1 //DTFT in system analysis
2 a=0.5; b=1;
3 n=0:50;
4 h=b*(a^n);
5 // Discrete-Time Fourier transform
6 \text{ K} = 500;
7 k=0:1:K;
8 \quad w = \%pi * k/K;
9 H=h*exp(-\%i*n'*w);
10 //x[n] is given as (a)^n*u[n]
11 xw=h*exp(-\%i*n'*w);
12 for i=1:501
       y(i) = H(i) *xw(i);
13
14 end
15 [phase_y,m]=phasemag(y);
```

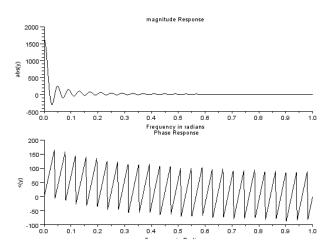


Figure 5.14: The DTFT in System Analysis

```
16 y=real(y);
17 subplot(2,1,1)
18 plot2d(w/%pi,y);
19 xlabel('Frequency in radians')
20 ylabel('abs(y)')
21 title('magnitude Response')
22 subplot(2,1,2)
23 plot2d(w/%pi,phase_y)
24 xlabel('Frequency in Radians');
25 ylabel('<(y)')
26 title('Phase Response')</pre>
```

Scilab code Exa 5.8b The DTFT in System Analysis

```
1 a=0.5; b=1;
2 n=0:50;
```

```
3 h=4*(a^n);
4 // Discrete-Time Fourier transform
5 \text{ K} = 500;
6 \text{ k=0:1:K};
7 w = \%pi * k/K;
8 H=h*exp(-%i*n'*w);
9 //x[n] is given as (a)^n*u[n]
10 x=4*[ones(1:51)];
11 xw = x * exp(%i*n'*w);
12 for i=1:501
       y(i) = H(i) * xw(i);
13
14 end
15 [phase_y,m]=phasemag(y);
16 y=real(y);
17 subplot(2,1,1);
18 plot2d(w/%pi,y);
19 xlabel('Frequency in radians')
20 ylabel('abs(y)')
21 title ('magnitude Response')
22 subplot (2,1,2)
23 plot2d(w/%pi,phase_y)
24 xlabel('Frequency in Radians');
25 ylabel('<(y)')
26 title('Phase Response')
```

### Scilab code Exa 5.9a DTFT and steady state response

```
1 //DTFT and steady state response
2 a=0.5,b=1;F=0.25;
3 n=0:(5/1000):5;
4 h=(a^n);
5 x=10*cos(0.5*%pi*n'+%pi/3);
6 H=h*exp(-%i*n'*F);
```

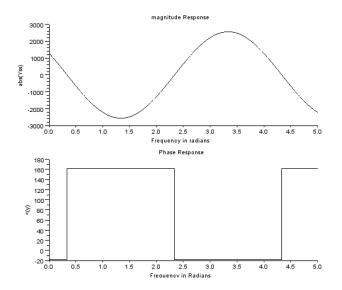


Figure 5.15: DTFT and steady state response

```
7 Yss=H*x;
8 [phase_Yss,m]=phasemag(Yss);
9 Yss=real(Yss);
10 subplot(2,1,1)
11 plot2d(n,Yss);
12 xlabel('Frequency in radians')
13 ylabel('abs(Yss)')
14 title('magnitude Response')
15 subplot(2,1,2)
16 plot2d(n,phase_Yss)
17 xlabel('Frequency in Radians');
18 ylabel('<(y)')
19 title('Phase Response')</pre>
```

Scilab code Exa 5.9b DTFT and steady state response

```
1 //DTFT and steady state response
2 a=0.8,b=1;F=0;
```

```
3 n=0:50;
4 h=(a^n);
5 x=4*[ones(1:10)];
6 H=h*exp(-%i*n'*F)
7 Yss=H*x
```

#### Scilab code Exa 5.10a System Representation in various forms

```
1 //System Representation in various forms
2 a=0.8; b=2;
3 n=0:50;
4 h=b*(a^n);
5 // Discrete-Time Fourier transform
6 \text{ K} = 500;
7 k=0:1:K;
8 \text{ w=\%pi*k/K};
9 H=h*exp(-\%i*n'*w);
10 // caluculation of phase and magnitude of h(z)
11 [phase_H,m]=phasemag(H);
12 H = abs(H);
13 subplot(2,1,1);
14 plot2d(w/%pi,H);
15 xlabel ('Frequency in radians')
16 ylabel('abs(H)')
17 title ('magnitude Response')
18 subplot(2,1,2)
19 plot2d(w/%pi,phase_H)
20 xlabel('Frequency in Radians');
21 ylabel('<(H)')
22 title('Phase Response')
```

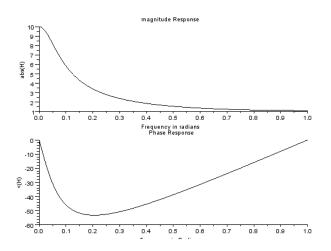


Figure 5.16: System Representation in various forms

Scilab code Exa 5.10b System Representation in various forms

```
1 //System Representation in various forms
2 a=0.6; b=1;
3 n=0:50;
4 h=b*(a^n);
5 // Discrete-Time Fourier transform
6 \text{ K} = 500;
7 k=0:1:K;
8 \quad w = \%pi * k/K;
9 H=h*exp(-\%i*n'*w);
10 //caluculation of phase and magnitude of h(z)
11 [phase_H,m]=phasemag(H);
12 \text{ H=abs}(H);
13 subplot(2,1,1);
14 plot2d(w/%pi,H);
15 xlabel('Frequency in radians')
16 ylabel('abs(H)')
17 title ('magnitude Response')
18 subplot (2,1,2)
```

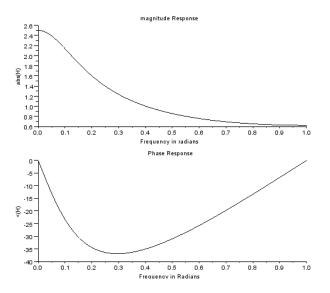


Figure 5.17: System Representation in various forms

```
19 plot2d(w/%pi,phase_H)
20 xlabel('Frequency in Radians');
21 ylabel('<(H)')
22 title('Phase Response')</pre>
```

## Chapter 6

## Filter Concepts

#### Scilab code Exa 6.1 The Minimum Phase Concept

```
1 //The minimum phase concept
2 z = \%z;
3 F=0:(0.5/400):0.5;
4 z = \exp(\%i * 2 * \%pi * F);
5 \text{ for } i=1:401
6 H1Z(i) = ((z(i)-1/2)*(z(i)-1/4))/((z(i)-1/3)*(z(i)
      -1/5));
7 end
8 \text{ for } i=1:401
9 H2Z(i) = (((-1/2)*z(i)+1)*(z(i)-1/4))/((z(i)-1/3)*(z(i)-1/3))
      )-1/5));
10 \, \text{end}
11 for i=1:401
12 H3Z(i) = (((-1/2)*z(i)+1)*((-1/4)*z(i)+1))/((z(i)-1/3)
      *(z(i)-1/5));
13 end
14 [phase_H1Z,m]=phasemag(H1Z);
15 [phase_H2Z,m]=phasemag(H2Z);
16 [phase_H3Z,m]=phasemag(H3Z);
```

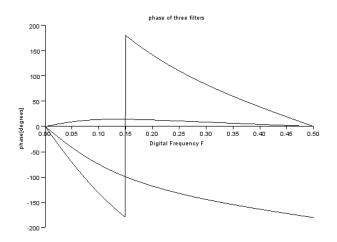


Figure 6.1: The Minimum Phase Concept

```
17 a=gca();
18 a.x_location="origin";
19 xlabel('Digital Frequency F');
20 ylabel('phase[degrees]');
21 xtitle('phase of three filters');
22 plot2d(F,phase_H1Z,rect=[0,-200,0.5,200]);
23 plot2d(F,phase_H2Z,rect=[0,-200,0.5,200]);
24 plot2d(F,phase_H3Z,rect=[0,-200,0.5,200]);
```

#### Scilab code Exa 6.4 Linear Phase Filters

```
1 //linear phase filters
2 z=%z
3 H1Z=((z^3)+2*(z^2)+2*z+1)/(z^3);
4 //from pole zero diagram its not a linear phase filter
5 H2Z=(z^4+4.25*z^2+1)/(z^4);
6 xset('window',1);
```

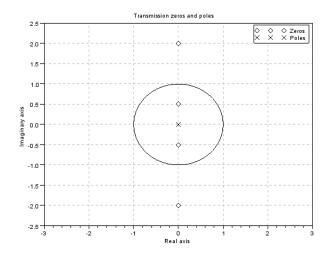


Figure 6.2: Linear Phase Filters

```
7 plzr(H2Z);
8 //from pole zero diagram and LPF
9 // characteristics its a linear phase filter
10 H3Z=((z^4+2.5*z^3-2.5*z-1)/(z^4));
11 xset('window',2);
12 plzr(H3Z);
13 //from pole zero diagram and LPF
14 // characteristics its a linear phase filter
```

Scilab code Exa 6.6 Frequency Response and Filter characteristics

```
1 //Frequency Response and filter characteristics
```

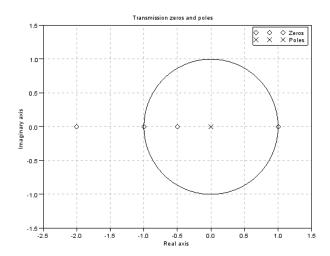


Figure 6.3: Linear Phase Filters

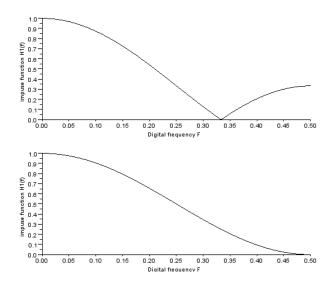


Figure 6.4: Frequency Response and Filter characteristics

```
2 z = \%z;
3 F=0:(0.5/200):0.5;
4 z = \exp(\%i * 2 * \%pi * F);
5 H1=(1/3)*(z+1+z^-1);
6 H2=(z/4)+(1/2)+(1/4)*(z^{-1});
7 \text{ H1} = abs(H1);
8 \text{ H2} = abs(H2);
9 a = gca();
10 a.x_location="origin";
11 subplot (211);
12 plot2d(F,H1);
13 xlabel('Digital frequency F');
14 ylabel('impuse function H1(f)');
15 subplot (212);
16 plot2d(F, H2);
17 xlabel('Digital frequency F');
18 ylabel('impuse function H1(f)');
```

#### Scilab code Exa 6.7a Filters and Pole Zero Plots

```
1 z=%z;
2 s=%s;
3 F=0:(0.5/400):0.5;
4 s=exp(%i*2*%pi*F);
5 H1Z=(z^4+1)/(z^4+1.6982*z^2+0.7210);
6 for i=1:401
7     H1(i)=(s(i)^4+1)/(s(i)^4+1.6982*s(i)^2+0.7210);
8 end
9 H1=abs(H1);
10 plzr(H1Z);
11 a=gca();
12 xset('window',1);
13 a.x_location="origin";
14 a.y_location="origin";
15 plot2d(F,H1)
```

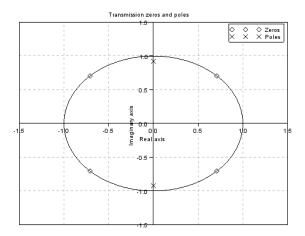


Figure 6.5: Filters and Pole Zero Plots

```
16 xlabel('Digital frequency F');
17 ylabel('magnitude');
18 xtitle('Magnitude spectrum of bandpass filter');
```

#### Scilab code Exa 6.7b Filters and Pole Zero Plots

```
1 z=%z;
2 s=%s;
3 F=0:(0.5/400):0.5;
4 s=exp(%i*2*%pi*F);
5 H1Z=(z^2+1-0.618*z)/(z^2-0.5857*z+0.898);
6 for i=1:401
7 H1(i)=(s(i)^2+1-0.618*s(i))/(s(i)^2-0.5857*s(i)+0.898);
```

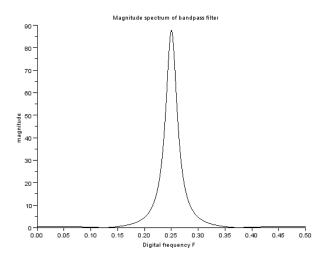


Figure 6.6: Filters and Pole Zero Plots

```
8 end
9 H1=abs(H1);
10 plzr(H1Z);
11 a=gca();
12 xset('window',1);
13 a.x_location="origin";
14 a.y_location="origin";
15 plot2d(F,H1)
16 xlabel('Digital frequency F');
17 ylabel('magnitude');
18 xtitle('Magnitude spectrum of bandpass filter');
```

Scilab code Exa 6.8 Digital resonator Design

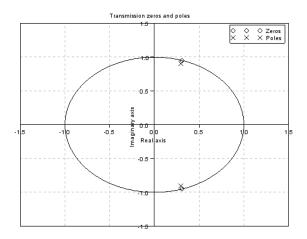


Figure 6.7: Filters and Pole Zero Plots

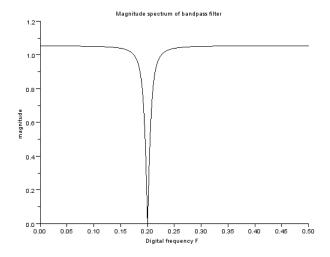


Figure 6.8: Filters and Pole Zero Plots

```
1 // Digital Resonator design with peak gain 50 HZ
2 //and 3 db bandwidth of 6HZ at sampling of 300 HZ
3 clf();
4 s = %s;
5 F = 0:150;
6 \text{ f=F/300};
7 s = \exp(\%i * 2 * \%pi * f);
8 for i=1:151
       H1(i) = (0.1054*(s(i)^2))/(s(i)^2-0.9372*s(i)
          +0.8783);
10 \text{ end}
11 H1 = abs(H1);
12 \text{ H2=H1}(40:60);
13 F1=40:60;
14 f1=F1/300;
15 a=gca();
16 a.x_location="origin";
17 a.y_location="origin";
18 plot2d(F,H1)
19 xlabel('Analog frequency F');
20 ylabel('magnitude');
21 xtitle ('Magnitude spectrum of digital resonator with
       peak 50HZ');
22 xset ('window',1);
23 a.x_location="origin";
24 a.y_location="origin";
25 plot2d(F1,H2)
26 xlabel('Analog frequency F');
27 ylabel('magnitude');
28 xtitle('passband detail');
```

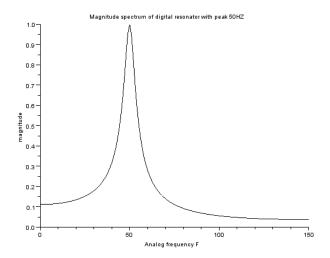


Figure 6.9: Digital resonator Design

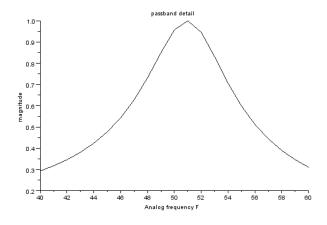


Figure 6.10: Digital resonator Design

#### Scilab code Exa 6.9 Periodic Notch Filter Design

```
1 // Periodic notch filter design at 60 HZ and sampling
       frequency 300HZ
2 z = \%z;
3 f=0:(0.5/400):0.5;
4 z1 = \exp(\%i * 2 * \%pi * f);
5 \text{ for } i=1:401
       H1Z(i) = (z1(i)^5-1)/((z1(i)^5)-(0.9^5));
       H2Z(i)=(z1(i)^5-1)/((z1(i)^5)-(0.99^5));
8 end
9 H1Z=abs(H1Z);
10 H2Z = abs(H2Z);
11 N1z=(1-z^-5)/(1-z^-1);
12 H3z=(N1z)/(horner(N1z,z/0.9));
13 H4z=(N1z)/(horner(N1z,z/0.99));
14 H3z=horner(H3z,z1);
15 H4z=horner(H4z,z1);
16 \quad a = gca();
17 a.x_location="origin";
18 a.y_location="origin";
19 plot2d(f,H1Z);
20 plot2d(f, H2Z);
21 xlabel('Digital frequency f');
22 ylabel ('magnitude');
23 xtitle ('Periodic Notch Filter N=5,R=0.9,0.99');
24 xset('window',1);
25 plot2d(f, H3z);
26 plot2d(f, H4z);
27 xlabel('Digital frequency f');
28 ylabel('magnitude');
29 xtitle ('Notch Filter that also passes DC N=5,R
      =0.9,0.99');
```

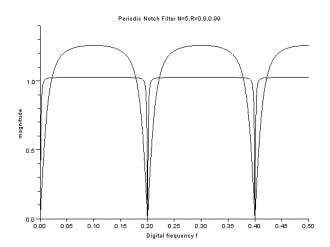


Figure 6.11: Periodic Notch Filter Design

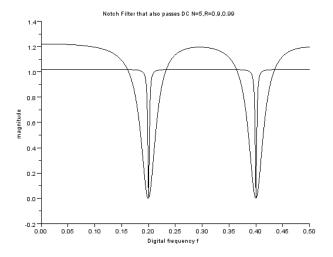


Figure 6.12: Periodic Notch Filter Design

### Chapter 7

## Digital Processing of Analog Signals

#### Scilab code Exa 7.3 Sampling oscilloscope

```
1 //Sampling Oscilloscope Concepts
2 \text{ fo=100; a=50;}
3 s=(a-1)*fo/a;
4 B = 100 - s;
5 i=s/(2*B);
6 i=ceil(i);
7 disp(i, 'The sampling frequency can at max divided by
8 disp(s,2*B, 'range of sampling rate is between s and
      2*B');
9 \text{ fol} = 100;
10 \ a=50;
11 s1=(a-1)*fo1/a;
12 B1=400-4*s1;
13 j=s1/(2*B1);
14 j=ceil(j);
15 disp(j, 'The sampling frequency can at max divided by
       j ');
16 disp(s1,2*B1, 'range of sampling rate is between s1
```

```
and 2*B1');
```

#### Scilab code Exa 7.4 Sampling of Band pass signals

```
1 //sampling of bandpass signals
2 \text{ fc=4;fl=6;}
3 B=f1-fc;
4 xt = [0 1 2 1];
5 xtt=[0 1 2];
6 a=0; b=1; c=2;
7 xta=[xt];
8 \text{ xtb} = [0 \ 0 \ 2 \ 1 \ 0];
9 \text{ xtc} = [0 \ 0 \ 0 \ 2 \ 1 \ 0];
10 xt1=[xta xta xta];
11 xt2=[xtb xtb(length(xtb):-1:2) xtb(2:length(xtb))
      xtb(length(xtb):-1:2)];
12 xt3=[xtc(length(xtc):-1:2) xtc(3:length(xtc)) zeros
      (1:7) xtc(length(xtc):-1:2) xtc(3:length(xtc))];
13 f1=0:length(xt1)-1;
14 	ext{ } f2 = [0 	ext{ } 1 	ext{ } 1.001 	ext{ } 2:6 	ext{ } 6.001 	ext{ } 7 	ext{ } .001 	ext{ } 8:12 	ext{ } 12.001];
15 f3=[-10:-8 -7.99 -7:-6 -5.99 -5:6 6.01 7:8 8.01
      9:10];
16 subplot (211);
17 plot2d(f1,xt1);
18 xtitle("spectrum of signal sampled at 4KHZ");
19 subplot (212);
20 plot2d(f2,xt2);
21 xtitle("spectrum of signal sampled at 7KHZ");
22 xset('window',1);
23 b=gca();
24 b.y_location="origin"
25 plot2d(f3,xt3);
26 xtitle("spectrum of signal sampled at 14KHZ");
```

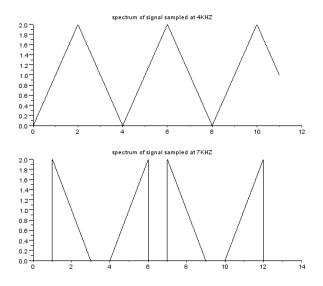


Figure 7.1: Sampling of Band pass signals

#### Scilab code Exa 7.6 Signal Reconstruction from Samples

```
1 //signal reconstruction from samples
2 //(a)By step interpolation method
3 x=[-1 2 3 2];
4 t=2.5;
5 ts=1;
6 t1=ceil(t);
7 t2=floor(t);
8 x1t=x(t2)
9 //(b)By linear interpolation method
10 x2t=(x(t1)+x(t2))/2
11 //(c)By sinc interpolation method
```

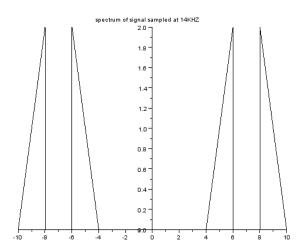


Figure 7.2: Sampling of Band pass signals

```
12 x3t=0; x1=[1 2 3 4];
13 for k=1:4
       x3t=x3t+(x1(k)*sinc(%pi*(t-(k-1))));
14
15 end
16 x3t//sinc interpolated value of x(2.5)
17 //(d)raised cosine interpolation method
18 \text{ x4t=0};
19 for k=1:4
       p = (\cos (0.5*\%pi*(t-k+1))/(1-(t-k+1)^2));
20
       xt=x1(k)*sinc(%pi*(t-k+1))*p;
21
22
       x4t = x4t + xt;
23 end
24 x4t//raised cosine interpolated value of x(2.5)
```

Scilab code Exa 7.7 Zero Interpolation and Spectrum Replication

```
1 //Zero interpolation and spectrum replication
2 XF=[0 1 2 1];
```

```
3 \text{ X1F} = [XF XF XF O];
4 YF=[X1F X1F];
5 \text{ DF=0.5*[XF XF 0]};
6 \text{ GF} = 0.5 * [XF \ 0 \ XF \ 0 \ XF \ 0];
7 f = -0.2:0.1:1;
8 f1 = -0.1:0.05:1.15;
9 f2=-0.4:0.2:1.2;
10 f3 = -0.2:0.1:1.2;
11 length(f3),length(GF)
12 a=gca();
13 a.y_location="origin";
14 subplot (211);
15 plot2d(f, X1F);
16 ylabel('X1F');
17 subplot (212);
18 a.y_location="origin";
19 plot2d(f1,YF);
20 ylabel('YF');
21 xset('window',1);
22 b=gca();
23 b.y_location="origin";
24 subplot (211);
25 plot2d(f2,DF);
26 ylabel('DF');
27 subplot (212);
28 b.y_location="origin";
29 plot2d(f3,GF);
30 ylabel('GF');
```

Scilab code Exa 7.8 Up Sampling and Filtering

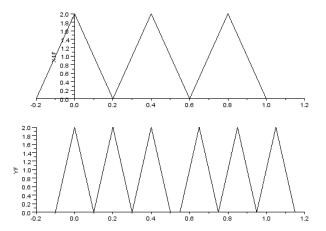


Figure 7.3: Zero Interpolation and Spectrum Replication

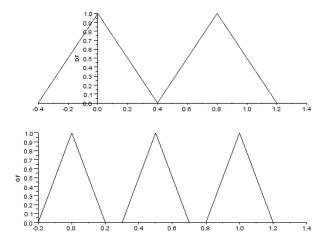


Figure 7.4: Zero Interpolation and Spectrum Replication

```
1 clf();
2 X = [0 0.5 1 0.5];
3 \text{ XF} = [X \text{ O}];
4 \text{ WF} = [X X X O];
5 f = -0.5:0.25:0.5;
6 f1 = -0.75:0.125:0.75;
7 \text{ HF} = [0 \ 1 \ 1 \ 1 \ 0];
8 f2=[-0.126, -0.125:0.125:0.125, 0.126];
9 \text{ for } i=1:5
        YF(i) = WF(i) * HF(i);
10
11 end
12 f3=[-0.126 -0.125 \ 0 \ 0.125 \ 0.126];
13 a=gca();
14 a.y_location="origin";
15 subplot (211);
16 plot2d(f,XF);
17 xtitle('spectrum of XF');
18 a.y_location="origin";
19 subplot (212);
20 plot2d(f1,WF);
21 xtitle('spectrum of WF');
22 xset('window',1);
23 b=gca();
24 b.y_location="origin";
25 subplot (211);
26 plot2d(f2, HF);
27 xtitle('spectrum of HF');
28 b.y_location="origin";
29 subplot (212);
30 plot2d(f3,YF);
31 xtitle('spectrum of YF');
```

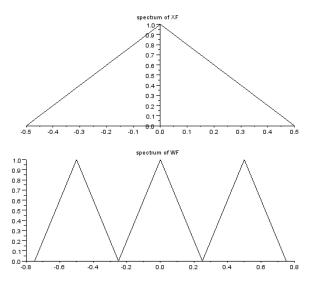


Figure 7.5: Up Sampling and Filtering

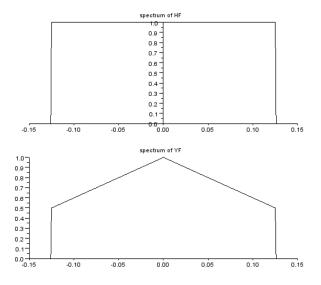


Figure 7.6: Up Sampling and Filtering

#### Scilab code Exa 7.9 Quantisation Effects

```
1 //(a) Quantisation effects
2 \text{ sig=0.005};
3 D=4;
4 B=log2(D/(sig*sqrt(12)));//no.of samples
5 //value of B to ensure quantisation error to 5mv
6 //(b) Quantisation error and noise
7 \text{ xn} = 0:0.2:2.0;
8 xqn=[0 \ 0 \ 0.5 \ 0.5 \ 1 \ 1 \ 1 \ 1.5 \ 1.5 \ 2 \ 2];
9 en=xn-xqn; // quantization error
10 // Quantisation signal top noise ratio
11 x=0; e=0;
12 for i=1:length(xn)
       x=x+xn(i)^2;
13
        e=e+en(i)^2;
14
15 end
16 / \text{method } 1
17 SNRQ = 10 * log 10 (x/e)
18 / \text{method } 2
19 SNRQ = 10 * log 10 (x/length (xn)) + 10.8 + 20 * log 10 (4) - 20 *
      log10(2)
20 SNRS=10*log10(1.33)+10*log10(12)+20*log10(4)-20*
      log10(2)
21 //from results we see that SNRS is statistical
      estimate
```

#### Scilab code Exa 7.10 ADC considerations

```
1 //ADC considerations
2 //(a) Aperture time TA
3 B=12;
4 fo=15000; //band limited frquency
5 TAm=(1/((2^B)))/(%pi*fo);
6 TAm=TAm*10^9
```

```
7 // Hence TA must satisfy TA<=TAm nano sec
8 //(b) conversion time of quantizer
9 TA = 4 * 10^{-9};
10 TH=10*10^-6; //hold time
11 S=30*10^3;
12 TCm = 1/S - TA - TH;
13 TCm=TCm * 10^6
14 //Hence TC must satisfy TC<=TCm micro sec
15 //(c) Holding capacitance C
16 \text{ Vo} = 10;
17 TH = 10 * 10^{-6};
18 B=12;
19 R=10^6; //input resistance
20 \text{ delv=Vo/(2^(B+1))};
21 Cm = (Vo*TH)/(R*delv);
22 \quad Cm = Cm * 10^9
23 //Hence C must satisfy C>=Cm nano farad
```

#### Scilab code Exa 7.11 Anti Aliasing Filter Considerations

```
//Anti Aliasing filter considerations
//minimum stop band attenuation As
B=input('enter no. of bits');//no. of samples
n=input('enter band width in KHZ');
As=20*log10(2^B*sqrt(6))
//nomalised frequency
Vs=(10^(0.1*As)-1)^(1/(2*n))
fp=4;//pass edge frequency
fs=Vs*fp//stop band frquency
S=2*fs//sampling frequency
fa=S-fp//aliaed frequency
Va=fa/fp;
//Attenuation at aliased frequency
Aa=10*log10(1+Va^(2*n))
```

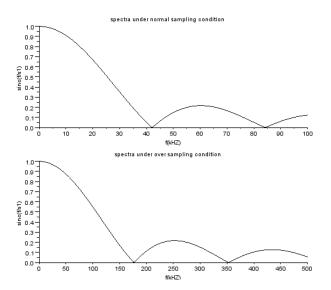


Figure 7.7: Anti Imaging Filter Considerations

#### Scilab code Exa 7.12 Anti Imaging Filter Considerations

```
//Anti Imaging Filter considerations
Ap=0.5;//passband attenuation
fp=20;//passband edge frequency
As=60;//stopband attenuation
S=42.1;
fs=S-fp;//stopband edge frequency
e=sqrt(10^(0.1*Ap)-1);
e1=sqrt(10^(0.1*Ap)-1);
n=(log10(e1/e))/(log10(fs/fp));
n=ceil(n)//design of nth order butworth filter
//(b) Assuming Zero-order hold sampling
S1=176.4;
fs1=S1-fp;
```

```
14 Ap=0.316;
15 e2=sqrt(10^{(0.1*Ap)-1)};
16 n1=(log10(e1/e2))/(log(fs1/fp));//new order of
     butworth filter
17 n1=ceil(n1)
18 f=0:100;
19 x = abs(sinc(f*\%pi/S));
20 f1=0:500;
21 x1=abs(sinc(f1*%pi/S1));
22 a=gca();
23 subplot (211);
24 plot2d(f,x);
25 xtitle("spectra under normal sampling condition", "f(
     kHZ)", "sinc(f/s1)");
26 subplot (212);
27 plot2d(f1,x1);
28 xtitle("spectra under over sampling condition", "f(
     kHZ)", "sinc(f/s1)");
```

## Chapter 8

# The Discrete Fourier Transform and its Applications

Scilab code Exa 8.1 DFT from Defining Relation

```
//DFT from defining relation
//N-point DFT
x=[1 2 1 0];
XDFT=dft(x,-1);
disp(XDFT, 'The DFT of x[n] is');
//DFT of periodic signal x with period N=4
```

Scilab code Exa 8.2 The DFT and conjugate Symmetry

```
1 //The DTFT and conjugate symmetry
2 //8-point DFT
3 x=[1 1 0 0 0 0 0 0];
4 XDFT=dft(x,-1);
5 disp(XDFT, 'The DFT of x is ');
6 disp('from conjugate symmetry we see XDFT[k]=XDFT[8-k]');
```

#### Scilab code Exa 8.3 Circular Shift and Flipping

```
1 // Circular shift and flipping
2 //(a) right circular shift
3 y = [1 2 3 4 5 0 0 6];
4 f = y; g = y; h = y;
5 \text{ for } i=1:2
       b=f(length(f));
6
7
       for j = length(f) : -1 : 2
8
            f(j)=f(j-1);
9
        end
10
        f(1) = b;
12 disp(f, 'By right circular shift y[n-2] is');
13 //(b) left circular shift
14 for i=1:2
15
       a=g(1);
       for j=1:length(g)-1
16
17
            g(j)=g(j+1);
18
        g(length(g))=a;
19
20 end
21 disp(g, 'By left circular shift y[n+2] is');
22 //(c) flipping property
23 h=[h(1) h(length(h):-1:2)];
24 disp(h, 'By flipping property y[-n] is');
```

#### Scilab code Exa 8.4 Properties of DFT

```
1 x=[1 2 1 0];
2 XDFT=dft(x,-1)
3 //(a)time shift property
```

```
4 y = x;
5 for i=1:2
       a = y(1);
7
       for j=1:length(y)-1
            y(j) = y(j+1);
9
        end
10
        y(length(y))=a;
11 end
12 YDFT = dft(y, -1)
13 disp(YDFT, 'By Time-Shift property DFT of x[n-2] is')
14 //(b) flipping property
15 g=[x(1) x(length(x):-1:2)]
16 GDFT = dft(g, -1)
17 disp(GDFT, 'By Time reversal property DFT of x[-n] is
18 //(c) conjugation property
19 p = XDFT;
20 PDFT = [p(1); p(4:-1:2)];
21 disp(YDFT, 'BY conjugation property DFT of x*[n] is')
```

#### Scilab code Exa 8.5a Properties of DFT

```
1 //properties of DFT
2 //a1)product
3 xn=[1 2 1 0];
4 XDFT=dft(xn,-1)
5 hn=xn.*xn
6 HDFT=dft(hn,-1)
7 HDFT1=1/4*(convol(XDFT,XDFT))
8 HDFT1=[HDFT1,zeros(8:12)];
9 HDFT2=[HDFT1(1:4);HDFT1(5:8);HDFT1(9:12)];
10 HDFT3=[0 0 0 0];
11 for i=1:4
```

```
12
        for j=1:3
13
             HDFT3(i) = HDFT3(i) + HDFT2(j,i);
14
        end
15 end
16 disp(HDFT3, 'DFT of x[n]^2 is');
17 //a2) periodic convolution
18 vn = convol(xn, xn);
19 vn=[vn, zeros(8:12)];
20 vn = [vn(1:4); vn(5:8); vn(9:12)];
21 \text{ vn1} = [0 \ 0 \ 0 \ 0];
22 \quad for \quad i = 1:4
23
        for j=1:3
24
             vn1(i)=vn1(i)+vn(j,i);
25
        end
26 \, \text{end}
27 VDFT = dft(vn1, -1);
28 VDFT1=XDFT.*XDFT;
29 disp(VDFT1, 'DFT of x[n]*x[n] is ');
30 //a3) signal energy (parcewell's theorem)
31 \text{ xn}2=\text{xn}^2;
32 E=0;
33 for i=1:length(xn2)
        E=E+abs(xn2(i));
34
35 end
36 \times DFT2 = XDFT^2
37 \text{ E1=0};
38 for i=1:length(XDFT2)
39
        E1=E1+abs(XDFT2(i));
40 \text{ end}
41 E, (1/4)*E1;
42 disp(1/4*E1, 'The energy of the signal is');
```

#### Scilab code Exa 8.5b Properties of DFT

```
1 / b1) modulation
```

```
2 XDFT = [4 -2*\%i \ 0 \ 2*\%i];
3 xn=dft(XDFT,1)
4 for i=1:length(xn)
        zn(i)=xn(i)*%e^((%i*%pi*(i-1))/2);
6 end
7 disp(zn, 'The IDFT of XDFT[k-1] is');
8 ZDFT = [2*\%i 4 -2*\%i 0];
9 zn1=dft(ZDFT,1)
10 //b2) periodic convolution
11 HDFT=(convol(XDFT, XDFT))
12 HDFT = [HDFT, zeros (8:12)];
13 HDFT = [HDFT (1:4); HDFT (5:8); HDFT (9:12)];
14 \text{ HDFT1} = [0 \ 0 \ 0 \ 0];
15 \text{ for } i=1:4
16
        for j=1:3
17
             HDFT1(i) = HDFT1(i) + HDFT(j,i);
18
        end
19 end
20 HDFT1;
21 hn=dft(HDFT1,1)
22 \text{ hn1}=4*(xn.*xn);
23 disp(hn1, 'The IDFT of XDFT*XDFT is');
24 / b3) product
25 WDFT=XDFT.*XDFT;
26 \text{ wn} = \text{dft}(\text{WDFT}, 1)
27 wn1 = convol(xn, xn);
28 wn1=[wn1, zeros(8:12)];
29 wn1=[wn1(1:4); wn1(5:8); wn1(9:12)];
30 \text{ WN} = [0 \ 0 \ 0 \ 0];
31 \text{ for } i=1:4
32
        for j=1:3
33
             WN(i) = WN(i) + wn1(j,i);
34
        end
35 end
36 disp(WN, 'The IDFT of XDFT.XDFT is');
37 //b4) Central ordinates and signal Energy
38 E=0;
39 for i=1:length(xn)
```

#### Scilab code Exa 8.5c Properties of DFT

```
1 // Regular convolution
2 xn=[1 2 1 0];
3 yn=[1 2 1 0 0 0 0];
4 YDFT=dft(yn,-1)
5 SDFT=YDFT.*YDFT
6 sn=dft(SDFT,1)
7 sn1=convol(xn,xn)
```

#### Scilab code Exa 8.6 Signal and Spectrum Replication

```
1 //Signal and spectrum replication
2 \text{ xn} = [2 \ 3 \ 2 \ 1];
3 \text{ XDFT} = \text{dft}(xn, -1)
4 yn = [xn xn xn];
5 \text{ YDFT} = \frac{\text{dft}}{\text{(yn,-1)}}
6 YDFT1=3*[XDFT(1:1/3:length(XDFT))];
7 \text{ for } i=2:3
        YDFT1(i:3:length(YDFT1))=0;
9 end
10 YDFT1 (12:-1:11)=0;
11 disp(YDFT1, 'the DFT of x[n/3] is');
12 hn = [xn(1:1/3:length(xn))]
13 for i=2:3
        hn(i:3:length(hn))=0;
14
15 end
16 hn(12:-1:11)=0;
17 hn
```

#### Scilab code Exa 8.7 Relating DFT and DTFT

```
1 //relating DFT and IDFT
2 XDFT1=[4 -2*%i 0 2*%i];
3 xn1=dft(XDFT1,1);
4 disp(xn1,'The IDFT of XDFT1');
5 XDFT2=[12 -24*%i 0 4*%e^(%i*%pi/4) 0 4*%e^(-%i*%pi/4) 0 24*%i];
6 xn2=dft(XDFT2,1);
7 disp(xn2,'The IDFT of XDFT1');
```

#### Scilab code Exa 8.8 Relating DFT and DTFT

```
1 // Relating DFT and DTFT
2 xn=[1 2 1 0];
3 XDFT=dft(xn,-1);
4 // for F=k/4,k=0,1,2,3
5 for k=1:4
6     XF(k)=1+2*%e^(-%i*%pi*(k-1)/2)+%e^(-%i*%pi*(k-1));
7 end
8 XF,XDFT
9 disp(XF, 'The DFT of x[n] is ');
```

Scilab code Exa 8.9a The DFT and DFS of sinusoids

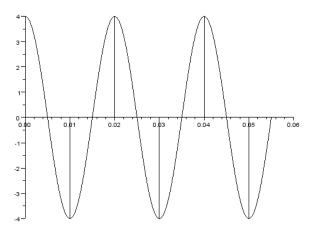


Figure 8.1: The DFT and DFS of sinusoids

```
1 //DFT and DFS of sinusoids
2 n2=0:0.5/1000:5.5/100;
3 xt=4*cos(100*%pi*n2');
4 n=0:(0.5)/100:(5.5)/100;//F=3/12 hence N=12
5 xn=4*cos(100*%pi*n');
6 XDFT=dft(xn,-1);
7 n1=0:11;
8 a=gca();
9 a.x_location="origin";
10 plot2d(n2,xt);
11 plot2d3('gnn',n,xn);
12 xset('window',1);
13 b=gca();
14 b.x_location="origin";
15 plot2d3('gnn',n1,XDFT);
```

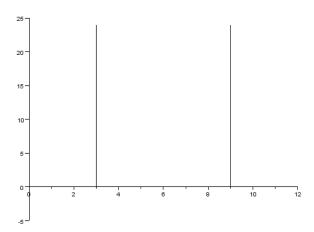


Figure 8.2: The DFT and DFS of sinusoids

#### Scilab code Exa 8.9b The DFT and DFS of sinusoids

```
1 //DFT and DFS of sinusoids
2 n2=0:1/1280:31/128;
3 xt=4*sin(72*%pi*n2');
4 n=0:1/128:31/128; //F=9/32 hence N=32
5 xn=4*sin(72*%pi*n');
6 XDFT=abs(dft(xn,-1));
7 n1=0:31;
8 a=gca();
9 a.x_location="origin";
10 plot2d(n2,xt);
11 plot2d3('gnn',n,xn);
12 xset('window',1);
13 b=gca();
14 b.x_location="origin";
15 plot2d3('gnn',n1,XDFT);
```

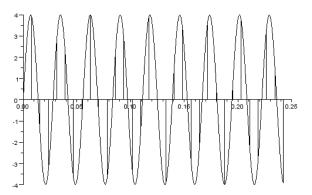


Figure 8.3: The DFT and DFS of sinusoids

#### Scilab code Exa 8.9c The DFT and DFS of sinusoids

```
1 //DFT and DFS of sinusoids
2 n2=0:1/840:6/21;
3 xt=4*sin(72*%pi*n2')-6*cos(12*%pi*n2');
4 n=0:1/21:6/21;//F=3/12 hence N=12
5 xn=4*sin(72*%pi*n')-6*cos(12*%pi*n');
6 XDFT=abs(dft(xn,-1));
7 n1=0:6;
8 a=gca();
9 a.x_location="origin";
10 plot2d(n2,xt);
11 plot2d3('gnn',n,xn);
```

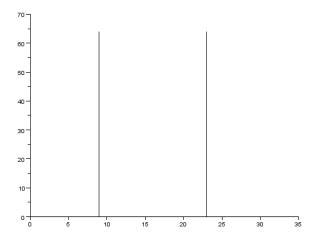


Figure 8.4: The DFT and DFS of sinusoids

## Scilab code Exa 8.9d The DFT and DFS of sinusoids

```
1 //DFT and DFS of sinusoids
2 n2=0:1/2400:23/240;
3 xt=1+4*sin(120*%pi*n2')+4*sin(40*%pi*n2');
4 n=0:1/240:23/240;//F=9/32 hence N=32
5 xn=1+4*sin(120*%pi*n')+4*sin(40*%pi*n');
6 XDFT=abs(dft(xn,-1));
7 n1=0:23;
```

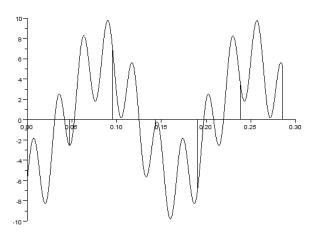


Figure 8.5: The DFT and DFS of sinusoids

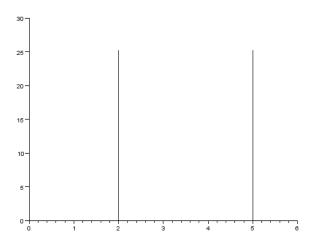


Figure 8.6: The DFT and DFS of sinusoids  $\,$ 

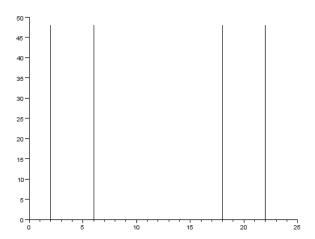


Figure 8.7: The DFT and DFS of sinusoids

```
8 a=gca();
9 a.x_location="origin";
10 plot2d(n2,xt);
11 plot2d3('gnn',n,xn);
12 xset('window',1);
13 b=gca();
14 b.x_location="origin";
15 plot2d3('gnn',n1,XDFT);
```

Scilab code Exa 8.10 DFS of sampled Periodic Signals

```
1 //DFS of sampled periodic signals
2 xn=[0 ones(2:16) 0 -ones(18:32)];
3 XDFS=0.032*dft(xn,-1);
```

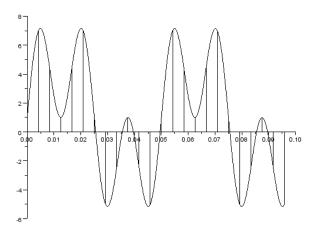


Figure 8.8: The DFT and DFS of sinusoids

```
4 for i=1:length(XDFS)
5     if (abs(XDFS(i)) < 0.000001) then
6         XDFS(i) = 0;
7     end
8 end
9 disp(XDFS, 'The DFS of x[n] is');</pre>
```

# Scilab code Exa 8.11 The effects of leakage

```
1 // Effects of leakage
2 n1=0:0.005:0.1;
3 n2=0:0.005:0.125;
4 n3=0:0.005:1.125;
5 xt1=(2*cos(20*%pi*n1')+5*cos(100*%pi*n1'));
6 xt2=(2*cos(20*%pi*n2')+5*cos(100*%pi*n2'));
7 xt3=(2*cos(20*%pi*n3')+5*cos(100*%pi*n3'));
8 XDFS1=abs(dft(xt1,-1))/20;
9 XDFS2=abs(dft(xt2,-1))/25;
```

```
10 XDFS3 = abs(dft(xt3,-1))/225;
11 f1=0:5:100;
12 f2=0:4:100;
13 f3=0:100/225:100;
14 a=gca();
15 a.x_location="origin";
16 plot2d3('gnn',f1,XDFS1);
17 xlabel('analog frequency');
18 ylabel ('Magnitude');
19 xset('window',1);
20 subplot (211);
21 plot2d3 ('gnn',f2, XDFS2);
22 xlabel('analog frequency');
23 ylabel('Magnitude');
24 subplot (212);
25 plot2d3('gnn',f3,XDFS3);
26 xlabel('analog frequency');
27 ylabel('Magnitude');
```

### Scilab code Exa 8.15a Methods to find convolution

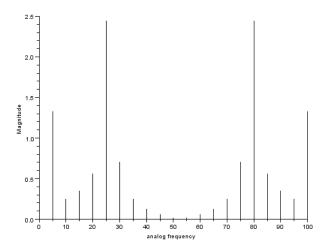


Figure 8.9: The effects of leakage

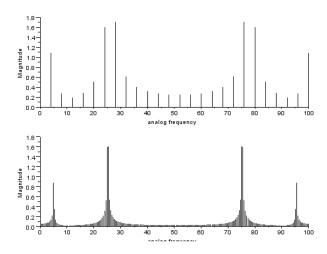


Figure 8.10: The effects of leakage

```
10 y1n=[0,0,0,y1n];

11 yn=yon+y1n

12 yn1=convol(xn,hn)
```

#### Scilab code Exa 8.15b Methods to find convolution

```
1 //(b) overlap -save method
2 xn=[1 2 3 3 4 5];
3 hn=[1 1 1];
4 xon=[0 0 1 2 3];
5 x1n=[2 3 3 4 5];
6 x2n=[4 5 0 0 0];
7 yon=convol(xon,hn);
8 y1n=convol(x1n,hn);
9 y2n=convol(x2n,hn);
10 yno=yon(3:5);
11 yn1=y1n(3:5);
12 yn2=y2n(3:5);
13 yn=[yno yn1 yn2]
14 YN=convol(xn,hn)
```

# Scilab code Exa 8.16 Signal Interpolation using FFT

```
1 //signal interpolation using FFT
2 xn=[0 1 0 -1];
3 XDFT=dft(xn,-1)
4 ZT=[0 -2*%i 0 zeros(1:27) 0 2*%i];
5 xn1=dft(ZT,1);
6 t=0:1/length(xn1):1-(1/length(xn1));
7 a=gca();
8 a.x_location="origin";
```

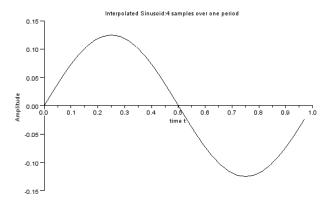


Figure 8.11: Signal Interpolation using FFT

```
9 plot2d(t,xn1);
10 xlabel('time t');
11 ylabel('Amplitude');
12 xtitle('Interpolated Sinusoid:4 samples over one period');
```

# Scilab code Exa 8.17 The Concept of Periodogram

```
1 //concept of periodogram
2 xn=[0 1 0 -1];
3 N=4;
4 XDFT=dft(xn,-1);
5 for i=1:length(XDFT)
6    p(i)=(1/N)*abs(XDFT(i)^2);
7 end
8 p//periodogram
```

## Scilab code Exa 8.18 DFT from matrix formulation

```
1 //The DFT from the matrix formulation
2 xn=[1;2;1;0];
3 w=exp(-%i*%pi/2);
4 for i=1:4
5     for j=1:4
6         WN(i,j)=w^((i-1)*(j-1));
7     end
8 end
9 XDFT=WN*xn
```

# Scilab code Exa 8.19 Using DFT to find IDFT

```
1 //using DFT to find IDFT
2 XDFT=[4;-2*%i;0;2*%i];
3 XDFTc=[4;2*%i;0;-2*%i];
4 w=exp(-%i*%pi/2);
5 for i=1:4
6     for j=1:4
7         WN(i,j)=w^((i-1)*(j-1));
8     end
9 end
10 xn=1/4*(WN*XDFTc)
```

Scilab code Exa 8.20 Decimation in Frequency FFT algorithm

```
1 //A four point decimation-in-frequency FFT algorithm
2 x=[1 2 1 0];
```

```
3  w=-%i;
4  xdft(1)=x(1)+x(3)+x(2)+x(4);
5  xdft(2)=x(1)-x(3)+w*(x(2)-x(4));
6  xdft(3)=x(1)+x(3)-x(2)-x(4);
7  xdft(4)=x(1)-x(3)-w*(x(2)-x(4));
8  XDFT=dft(x,-1);
9  xdft, XDFT
```

## Scilab code Exa 8.21 Decimation in time FFT algorithm

```
1 //A four point decimation—in—time FFT algorithm
2 x=[1 2 1 0];
3 w=-%i;
4 xdft=[0 0 0 0];
5 for i=1:4
6     for j=1:4
7         xdft(i)=xdft(i)+x(j)*w^((i-1)*(j-1));
8     end
9 end
10 XDFT=dft(x,-1);
11 xdft,XDFT
```

# Scilab code Exa 8.22 4 point DFT from 3 point sequence

```
1 //A 4-point DFT from a 3-point sequence
2 xn=[1;2;1];
3 w=exp(-%i*%pi/2);
4 for i=1:4
5     for j=1:3
6         WN(i,j)=w^((i-1)*(j-1));
7     end
8 end
9 XDFT=WN*xn
```

# Scilab code Exa 8.23 3 point IDFT from 4 point DFT

```
1 //A 3-point IDFT from 4-point DFT
2 XDFT=[4;-2*%i;0;2*%i];
3 w=exp(-%i*%pi/2);
4 for i=1:4
5     for j=1:3
6         WN(i,j)=w^((i-1)*(j-1));
7     end
8 end
9 WI=WN';
10 xn=1/4*(WI*XDFT)
```

## Scilab code Exa 8.24 The importance of Periodic Extension

```
1 //The importance of Periodic extension
\frac{2}{(a)} = \frac{1}{(a)} = 3
3 x = [1 2 1];
4 XDFT = dft(x, -1)
5 \text{ w=exp}(-\%i*2*\%pi/3);
6 \text{ for } i=1:3
         for j=1:3
              WN(i,j)=w^{((i-1)*(j-1))};
8
9
         end
10 \text{ end}
11 WI = WN';
12 \text{ xn} = 1/3 * WI * XDFT
13 //The result is periodic with M=3 & 1 period equals
       x[n]
14 //(b) For M=4
15 y = [1 \ 2 \ 1 \ 0];
```

```
16  YDFT=dft(y,-1)
17  w=exp(-%i*%pi/2);
18  for i=1:4
19         for j=1:4
20             WN(i,j)=w^((i-1)*(j-1));
21         end
22  end
23  WI=WN';
24  yn=1/4*WI*YDFT
```

# Chapter 9

# Design of IIR Filters

Scilab code Exa 9.1 Response Invariant Mappings

```
1 //Response in variant mappings
2 s = \%s; z = \%z;
3 \text{ HS}=1/(s+1);
4 f = 0:0.05:0.5;
5 HS1=horner(HS,(%i*%pi*2*f'));
6 \text{ ts=1};
7 HZ=z/(z-0.3679);
8 HZ1=horner(HZ, exp(%i*%pi*2*f'));
9 a = gca();
10 a.x_location="origin";
11 subplot (211)
12 plot2d(f, HS1);
13 plot2d(f, HZ1);
14 xlabel('Analog frequency f(Hz)');
15 ylabel('Magnitude');
16 xtitle('magnitude of H(s) and H(z)');
17 HZ1=HZ1-0.582; //magnitude after gain matching at dc
18 b=gca();
19 b.x_location="origin";
20 subplot (212);
21 plot2d(f, HS1);
```

```
22 plot2d(f, HZ1);
23 xlabel('Analog frequency f(Hz)');
24 ylabel('Magnitude');
25 xtitle('magnitude after gain matching at DC');
26 //Impulse response of analog and digital filter
27 t=0:0.01:6;
28 ht = exp(-t');
29 \quad n=0:6;
30 hn = exp(-n');
31 xset ('window',1)
32 c=gca();
33 subplot (211);
34 plot2d(t,ht);
35 plot2d3('gnn',n,hn);
36 xlabel('DT index n and time t=nts');
37 ylabel('Amplitude');
38 xtitle ('Impulse response of analog and digital
      filter');
39 //Step response of analog and digital filter
40 t=0:0.01:6;
41 st=1-exp(-t');
42 \quad n=0:6;
43 sn = (\%e - \%e^{(-n')})/(\%e - 1);
44 c=gca();
45 subplot (212);
46 plot2d(t,st);
47 plot2d3('gnn',n,sn);
48 xlabel('DT index n and time t=nts');
49 ylabel('Amplitude');
50 xtitle('Step response of analog and digital filter')
```

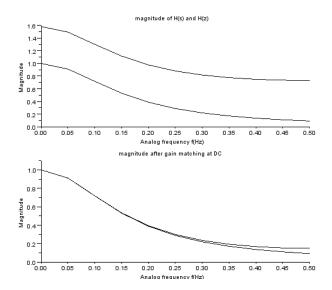


Figure 9.1: Response Invariant Mappings

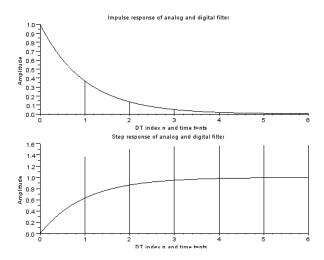


Figure 9.2: Response Invariant Mappings

### Scilab code Exa 9.2 Impulse Invariant Mappings

```
1 //Impulse invariant mappings
\frac{2}{\sqrt{a}} = \frac{1}{\sqrt{a}} = \frac{4s + 7}{s^2 + 5s + 4} to H(z) using
      impulse invariance
3 s = %s;
4 z = \%z;
5 HS = (4*s+7)/(s^2+5*s+4);
6 pfss(HS)
7 \text{ ts} = 0.5;
8 HZ=3*z/(z-\%e^{(-4*ts)})+z/(z-\%e^{(-ts)})
9 //(b) converting H(s)=4s+7/s^2+5s+4 to H(z) using
      impulse invariance
10 HS1=4/((s+1)*(s^2+4*s+5))
11 pfss(HS1);
12 HZ1=2*z/(z-\%e^-ts)+(2*1.414*z^2*\cos(-0.75*\%pi)
      -2*1.414*(z/\%e)*\cos(0.5-0.75*\%pi))/(z^2-2*(z/\%e)*
      \cos(0.5) + \%e^{-2}
```

## Scilab code Exa 9.3ab Modified Impulse Invariant Design

```
1 //(a) Impulse invariant design
2 s=%s;z=%z;
3 HS=1/(s+1);
4 H1s=horner(HS,3*s/%pi)
5 H1z=%pi/3*z/(z-%e^(-%pi/3))
6 // Modified inmpulse invariant design
7 HZ=z/(z-1/%e);
8 HMZ=0.5*(z+1/%e)/(z-1/%e);// modified transfer function
9 H1Z=HZ/horner(HZ,1)
10 HM1Z=HMZ/horner(HMZ,1)
11 f=0:0.05:0.5;
```

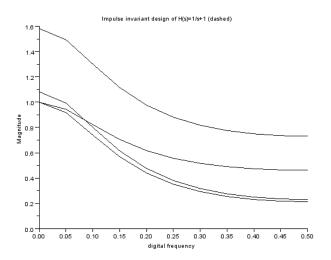


Figure 9.3: Modified Impulse Invariant Design

```
12  HZ1=horner(HZ,exp(%i*2*%pi*f'));
13  HMZ1=horner(HMZ,exp(%i*2*%pi*f'));
14  H1Z1=horner(H1Z,exp(%i*2*%pi*f'));
15  HM1Z1=horner(HM1Z,exp(%i*2*%pi*f'));
16  a=gca();
17  a.x_location="origin";
18  plot2d(f,HZ1);
19  plot2d(f,HMZ1);
20  plot2d(f,HMZ1);
21  plot2d(f,HMIZ1);
22  xlabel('digital frequency');
23  ylabel('Magnitude');
24  xtitle('Impulse invariant design of H(s)=1/s+1 (dashed)');
```

Scilab code Exa 9.3cd Modified Impulse Invariant Design

```
1 //modified Impulse invariant Design
```

```
\frac{2}{\sqrt{(c)}}H(s)=4s+7/s<sup>2</sup>+5s+4
3 s = \%s; z = \%z;
4 HS = (4*s+7)/(s^2+5*s+4);
5 [d1]=degree(numer(HS));
6 [d2]=degree(denom(HS));
7 HZ = ((3*z)/(z-\%e^2-2))+(z/(z-\%e^2-0.5))
8 \text{ if } (d2-d1==1) \text{ then}
9
        h = (4+7/\%inf)/(1+5/\%inf+4/\%inf)
        HMZ = HZ - 0.5 * h
10
11 else
12
        HMZ = HZ
13 end
14 HS1=4/((s+1)*(s^2+4*s+5))
15 HZ1 = (0.2146*z^2+0.093*z)/(z^3-1.2522*z^2+0.527*z
       -0.0821);
16 [d1]=degree(numer(HS1));
17 [d2] = degree (denom (HS1));
18 \text{ if } (d2-d1==1) \text{ then}
        HMZ1=HZ1-0.5*h
19
20 else
21
        HMZ1 = HZ1
22 end
```

#### Scilab code Exa 9.5 Mappings from Difference Algorithms

```
//Mappings from difference algorithms
//Backward difference mappings
s=%s;z=%z;
ts=1;a=1;
HS=1/(s+a);
HZa=horner(HS,(z-1)/(z*ts))
z1=roots(denom(HZa))//for ts >0 HZa always stable
HZb=horner(HS,(z-1)/ts)
z2=roots(denom(HZb))//stable only for 0<ats<2
HZc=horner(HS,(z^2-1)/(2*z*ts))</pre>
```

```
11 z3=roots(denom(HZc))//magnitude of 1 pole is always >1 hence unstable
```

## Scilab code Exa 9.6 Mappings From Integration Algorithms

## Scilab code Exa 9.7 DTFT of Numerical Algorithms

```
1 //DTFT of numerical algorithms
2 //(a)For trapezoidal numerical integrator
3 ieee(2)
4 F=0:0.01:0.2;
5 HTF=1/(%i*2*tan(%pi*F'));
6 HIF=1/(%i*2*%pi*F');
7 Ha=1-((%pi*F')^2)/3-((%pi*F')^4/45);
8 //(b)For simphson's numerical integrator
9 Hb=((2*%pi*F')).*((2+cos(2*%pi*F'))./(3*sin(2*%pi*F')));
10 //For forward difference operator
```

```
11 HFF = (\%e^{(\%i*2*\%pi*F')}) - 1;
12 HDF=1/(%i*2*%pi*F');
13 Hc=1+(%i*2*\%pi*F')/2-(2*\%pi*F')^2/6;
14 Hc = abs(Hc);
15 //for central difference operator
16 HCF=sin(2*%pi*F')./(%i*4*%pi*%pi*F',^2);
17 Hd=abs(sin(2*%pi*F')./(2*%pi*F'));
18 length(F), length(Ha)
19 a=gca();
20 a.x_location="origin";
21 plot2d(F, Ha, rect = [0, 0.8, 0.2, 1.1]);
22 plot2d(F, Hb);
23 xtitle ("Magnitude spectrum of Integration algorithms
      ", "Digital Frequency F", "Magnitude");
24 xset ('window', 1);
25 plot2d(F, Hc, rect = [0, 0.8, 0.2, 1.1]);
26 plot2d(F, Hd);
27 xtitle ("Magnitude spectrum of difference algorithms"
      "," Digital Frequency F", "Magnitude,");
```

#### Scilab code Exa 9.8a Bilinear Transformation

```
// Bilinear transformation
//To convert bessel analog filter to digital filter
s=%s;
z=%z;
HS=3/(s^2+3*s+3);
Wa=4;//analog omega
Wd=%pi/2;//digital omega
```

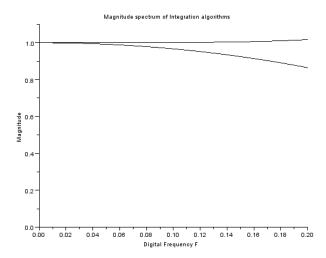


Figure 9.4: DTFT of Numerical Algorithms

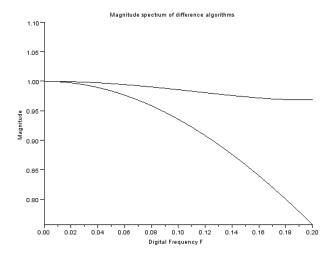


Figure 9.5: DTFT of Numerical Algorithms

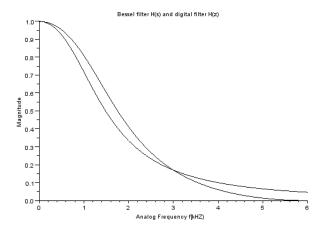


Figure 9.6: Bilinear Transformation

```
8 T=(2/Wa)*(tan(Wd/2));
9 HZ=horner(HS,(2/T)*(z-1)/(z+1))
10 f=0:0.1:6;
11 HS1=horner(HS,(%i*4*f'/3));
12 HS1=abs(HS1);
13 HZ1=horner(HZ,exp(-%i*%pi*f'/6));
14 HZ1=abs(HZ1);
15 a=gca();
16 a.x_location="origin";
17 plot2d(f,HS1);
18 plot2d(f,HS1);
19 xlabel('Analog Frequency f[kHZ)');
20 ylabel('Magnitude');
21 xtitle('Bessel filter H(s) and digital filter H(z)');
;
```

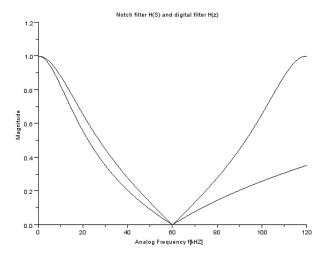


Figure 9.7: Bilinear Transformation

#### Scilab code Exa 9.8b Bilinear Transformation

```
1 //Bilinear transformation
2 //To convert twin-T notch analog filter to digital
      filter
3 s = %s;
4 z = %z;
5 HS=(s^2+1)/(s^2+4*s+1);
7 S=240; f=60; //sampling and analog frequencies
8 W=0.5*%pi;//digital frequency
9 C=Wo/tan(0.5*W)
10 HZ=horner(HS,C*(z-1)/(z+1))
11 f=0:120;
12 HZ1=abs(horner(HZ,exp(-%i*%pi*f'/120)));
13 HS1=abs(horner(HS,(%i*f'/60)));
14 a=gca();
15 a.x_location="origin";
16 plot2d(f, HZ1);
17 plot2d(f, HS1);
```

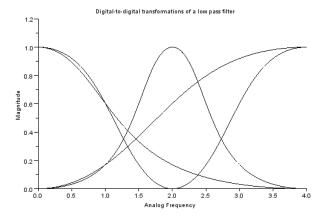


Figure 9.8: D2D transformations

```
18 xlabel('Analog Frequency f[kHZ]');
19 ylabel('Magnitude');
20 xtitle('Notch filter H(S) and digital filter H(z)');
```

#### Scilab code Exa 9.9 D2D transformations

```
10 W2=0.75*\%pi;
11 K=tan(0.5*WD)/tan(0.5*(W2-W1))
12 a = -\cos(0.5*(W1+W2))/\cos(0.5*(W1-W2))/Mapping
      parameters k, a, A1, A2
13 A1=2*K*a/(K+1), A2=(K-1)/(K+1)
14 HBPZ=horner(HZ, -(z^2+A1*z+A2)/(A2*z^2+A1*z+1))
15 //(c)Lp 2 BS transformation
16 w1=3/8*\%pi; w2=5/8*\%pi; //band edges
17 K1 = tan(0.5*WD)*tan(0.5*(w2-w1))
18 a1=-\cos(0.5*(w1+w2))/\cos(0.5*(w2-w1))//\text{Mapping}
      parameters k1, a1, A1c, A2c
19 A1c=2*a1/(K1+1), A2c=-(K1-1)/(K1+1)
20 HBSZ=horner(HZ,(z^2+A1c*z+A2c)/(A2c*z^2+A1c*z+1))
21 f = 0:1/256:4;
22 for i=1:length(f)
23 HZ1(i)=horner(HZ, exp(-\%i*\%pi*f(i)/4));
24 end
25 \text{ HZ1} = abs(HZ1);
26 HHPZ1=abs(horner(HHPZ, exp(-%i*\%pi*f'/4)));
27 HBPZ1=abs(horner(HBPZ,exp(-%i*%pi*f',/4)));
HBSZ1=abs(horner(HBSZ, exp(-\%i*\%pi*f',/4)));
29 \ a = gca();
30 a.x_location="origin";
31 plot2d(f, HZ1);
32 plot2d(f, HHPZ1);
33 plot2d(f, HBPZ1);
34 plot2d(f, HBSZ1);
35 xlabel('Analog Frequency');
36 ylabel('Magnitude');
37 xtitle ('Digital-to-digital transformations of a low
      pass filter');
```

Scilab code Exa 9.10a Bilinear Design of Second Order Filters

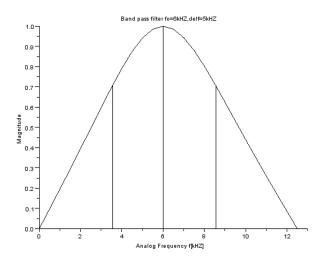


Figure 9.9: Bilinear Design of Second Order Filters

```
1 // Bilinear design of second order filters
2 s = \%s; z = \%z;
3 fo=6; Wo=2*%pi*fo/25;
4 delf=5;S=25;
5 B = \cos(2*\%pi*fo/25)
6 C = tan(\pi/25)
7 HS=1/(s+1);
8 HZ=horner(HS,(z^2-(2*B*z)+1)/(C*(z^2)-C))
9 f=0:0.5:12.5;
10 HZ1=horner(HZ, exp(%i*2*%pi*f'/25));
11 HZ1 = abs(HZ1);
12 W2=(\%pi*delf/25)+acos(cos(Wo)*cos(\%pi*delf/25))
13 W1=W2-(2*\%pi*delf/25)
14 f1=S*W1/(2*\%pi), f2=S*W2/(2*\%pi)
15 f3=[f1;fo;f2];
16 HZf=abs(horner(HZ,exp(%i*2*%pi*f3'/25)));
17 a=gca();
18 a.x_location="origin";
19 plot2d(f, HZ1, rect = [0 0 13 1]);
20 plot2d3('gnn',f3,HZf);
```

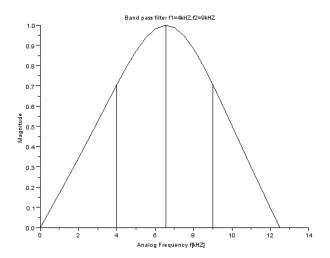


Figure 9.10: Bilinear Design of Second Order Filters

```
21 xlabel('Analog Frequency f[kHZ]');
22 ylabel('Magnitude');
23 xtitle('Band pass filter fo=6kHZ, delf=5kHZ');
```

# Scilab code Exa 9.10b Bilinear Design of Second Order Filters

```
1 // Bilinear design of second order filters
2 s=%s;z=%z;
3 f1=4;f2=9;
4 delf=f2-f1;S=25;
5 B=cos(%pi*(f1+f2)/25)/cos(%pi*(f2-f1)/25)
6 C=tan(%pi*delf/25)
7 HS=1/(s+1);
8 HZ=horner(HS,(z^2-(2*B*z)+1)/(C*(z^2)-C))
9 f=0:0.5:12.5;
10 HZ1=horner(HZ,exp(%i*2*%pi*f'/25));
```

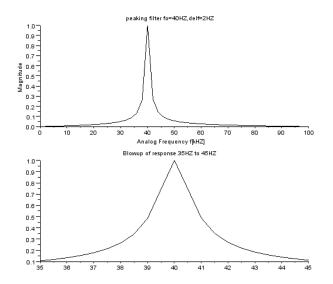


Figure 9.11: Bilinear Design of Second Order Filters

```
11 HZ1=abs(HZ1);
12 fo=S*acos(B)/(2*%pi)
13 f3=[f1 fo f2];
14 HZf=abs(horner(HZ,exp(-%i*2*%pi*f3'/25)));
15 a=gca();
16 a.x_location="origin";
17 plot2d(f,HZ1);
18 plot2d3('gnn',f3,HZf);
19 xlabel('Analog Frequency f[kHZ]');
20 ylabel('Magnitude');
21 xtitle('Band pass filter f1=4kHZ,f2=9kHZ');
```

Scilab code Exa 9.10c Bilinear Design of Second Order Filters

```
1 //Bilinear design of second order filters
2 s=%s;z=%z;
```

```
3 fo=40; Wo=2*\%pi*fo/200;
4 delf=2; S=25;
5 \text{ delW}=2*\%pi*delf/200;
6 B = \cos(2*\%pi*fo/200)
7 K=0.557;
8 \quad C=K*tan(0.5*delW)
9 HS=1/(s+1);
10 HZ=horner(HS,(z^2-(2*B*z)+1)/(C*(z^2)-C))
11 f = 0:2:100;
12 f1=35:0.5:45;
13 HZ1=horner(HZ, exp(%i*2*%pi*f', 200));
14 HZ2=horner(HZ, exp(%i*2*%pi*f1', 200));
15 HZ1 = abs(HZ1);
16 HZ2=abs(HZ2);
17 a = gca();
18 a.x_location="origin";
19 subplot (211);
20 plot2d(f, HZ1);
21 xlabel('Analog Frequency f[kHZ]');
22 ylabel('Magnitude');
23 xtitle('peaking filter fo=40HZ, delf=2HZ');
24 subplot (212);
25 plot2d(f1,HZ2);
26 xtitle('Blowup of response 35HZ to 45HZ');
```

## Scilab code Exa 9.11 Interference Rejection

```
//interference Rejection
//design oh high-Q and low-Q notch filters
s=%s;z=%z;
Q=50;
fo=60;S=300;
delf=fo/Q;
```

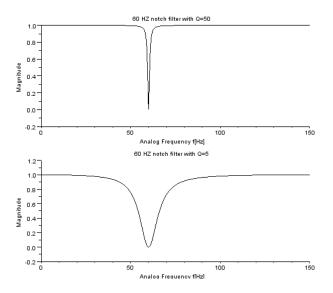


Figure 9.12: Interference Rejection

```
7 Wo=2*\%pi*fo/S;
8 delW=2*%pi*delf/S;
9 C=tan(0.5*delW), B=cos(Wo)
10 HS = (s)/(s+1);
11 H1Z = horner(HS, (z^2 - (2*B*z) + 1) / (C*(z^2) - C))
12 \quad Q1=5; delf1=fo/Q1;
13 delW1=2*\%pi*delf1/S;
14 C1 = tan(0.5 * delW1), B1 = cos(Wo)
15 H2Z=horner(HS,(z^2-(2*B1*z)+1)/(C1*(z^2)-C1))
16 \quad f = 0:0.5:150;
17 H1Z1=horner(H1Z,exp(%i*2*%pi*f',/S));
18 H2Z1=horner(H2Z, exp(%i*2*%pi*f',/S));
19 a=gca();
20 subplot (211);
21 plot2d(f, H1Z1);
22 xlabel('Analog Frequency f[Hz]');
23 ylabel('Magnitude');
24 xtitle('60 HZ notch filter with Q=50');
25 subplot (212);
26 plot2d(f, H2Z1);
```

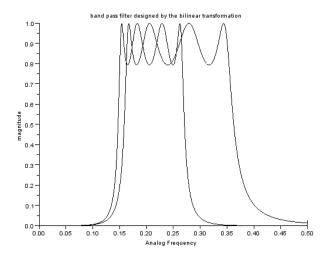


Figure 9.13: IIR Filter Design

```
27 xlabel('Analog Frequency f[Hz]');
28 ylabel('Magnitude');
29 xtitle('60 HZ notch filter with Q=5');
```

# Scilab code Exa 9.12 IIR Filter Design

```
9 omega=2*tan(0.5*W');//prewarping each band edge
                      frequency
10 epsilon=sqrt(10^(0.1*Ap)-1);
11 n=acosh(((10^(0.1*As)-1)/epsilon^2)^1/2)/(acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1/acosh(fs1
                     fp1));
12 n = ceil(n)
13 alpha=(1/n)*asinh(1/epsilon);
14 for i=1:n
                          B(i) = (2*i-1)*\%pi/(2*n);
15
16 end
17 \text{ for } i=1:n
                          p(i) = -\sinh(alpha) * \sin(B(i)) + \%i * \cosh(alpha) * \cos(B)
                                      (i));
19 end
20 \ Qs = 1;
21 for i=1:n
                           Qs=Qs*(s-p(i))
22
23 end
24 \quad Qo = 0.1634;
25 \text{ HPS=Qo/Qs}
26 HBPS=horner(HPS,(s^2+1.5045^2)/(s*1.202))
27 HZ=horner(HBPS, 2*(z-1)/(z+1))
28 \quad f = 0:0.001:0.5;
29 HZF=abs(horner(HZ,exp(%i*2*%pi*f')));
30 HBPF=abs(horner(HBPS, %i*2*%pi*f'));
31 a=gca();
32 plot2d(f, HZF);
33 plot2d(f, HBPF);
34 xlabel('Analog Frequency');
35 ylabel('magnitude');
36 xtitle ('band pass filter designed by the bilinear
                      transformation');
```

# Chapter 10

# Design of FIR filters

# Scilab code Exa 10.2 Truncation and Windowing

```
1 //Truncation and Windowing
2 //(a)N=9, Barlett Window.
3 z = \%z;
4 Fc = 0.25;
5 n = -4:4;
6 hn=2*Fc*(sinc(0.5*n',*%pi))
7 Wn=1-(2*abs(n'))/8//Barlett window
8 \text{ hwn=hn.*Wn}
9 Hcz=0;
10 for i=1:length(hwn)
         Hcz=Hcz+hwn(i)*(z^{(2-i))};
12 end
13 Hcz//indicates delay of 0.15ms
14 //(b)N=6, vonhann Window
15 n1 = -2.5:2.5;
16 hn1=2*Fc*(sinc(0.5*n1',*%pi))
17 Wn1=0.5+0.5*(cos(0.4*%pi*n1'))//Vonhann window
18 \, \text{hwn1} = \text{hn1}.* \text{Wn1}
19 Hcz1=0;
20 for i=1:length(hwn1)
         Hcz1=Hcz1+hwn1(i)*(z^{((2-i))});
```

```
22 end 23 Hcz1//1st sample of hwn is 0 hence delay is 1.5ms
```

### Scilab code Exa 10.3ab FIR lowpass Filter design

```
1 //FIR filter design using windows
2 //(a) Design of FIR filter to meet following
      specifications
3 fp=2; fs=4; Ap=2; As=40; S=20;
4 Fp=fp/S;Fs=fs/S;
5 Fc = 0.15;
6 z=\%z;
7 \text{ N1=3.21/(Fs-Fp)};
8 \text{ N1} = \text{ceil}(\text{N1})
9 N2=5.71/(Fs-Fp);
10 N2 = ceil(N2)
11 n1 = -16:16;
12 \quad n2 = -28.5:1:28.5;
13 hn1=2*Fc*(sinc(2*Fc*n1'));
14 hn2=2*Fc*(sinc(2*Fc*n2'));
15 Wn1=0.5+0.5*(cos(2*%pi*n1'/(N1-1)));//Vonhann window
16 \text{Wn2}=0.42+0.5*(\cos(2*\%\text{pi}*\text{n2}'/(\text{N2}-1)))+0.08*(\cos(4*\%\text{pi}))
      *n2', (N2-1))); //Blackman window
17 hwn1 = abs(hn1.*Wn1);
18 hwn2=abs(hn2.*Wn2);
19 [hwn1F, fr1] = frmag(hwn1, 256);
20 [hwn2F,fr2]=frmag(hwn2,256);
21 hwn1F1=20*log10(hwn1F);
22 hwn2F1=20*log10(hwn2F);
23 plot2d(fr1,hwn1F1);
24 plot2d(fr2(1:length(fr2)-2), hwn2F1(1:length(fr2)-2))
25 xlabel('Digital frequency');
26 ylabel('Magnitude [dB]');
27 title ('Low pass filter using vonhann and Blackmann
```

```
windows Fc = 0.15, vonhann N = 33, Blackman N = 58');
28 //(b) Minimum length design
29 \text{ Fcv} = 0.1313;
30 \text{ Fcb=0.1278};
31 \text{ Nv} = 23; \text{Nb} = 29;
32 \text{ nv} = -11:11;
33 \text{ nb} = -14:14;
34 \text{ hnv} = 2 * \text{Fcv} * (\text{sinc}(2 * \text{Fcv} * \text{nv}'));
35 hnb=2*Fcb*(sinc(2*Fcb*nb'));
36 \text{Wnv} = 0.5 + 0.5 * (\cos(2*\%\text{pi}*\text{nv}'/(\text{Nv}-1))); //\text{Vonhann window}
37 \text{Wnb}=0.42+0.5*(\cos(2*\%pi*nb'/(Nb-1)))+0.08*(\cos(4*\%pi))
       *nb'/(Nb-1)));//Blackman window
38 hwnv=abs(hnv.*Wnv);
39 \text{ hwnb=abs}(hnb.*Wnb);
40 [hwnvF,frv]=frmag(hwnv,256);
41 [hwnbF,frb]=frmag(hwnb,256);
42 hwnvF = 20 * log 10 (hwnvF);
43 hwnbF=20*log10(hwnbF);
44 b=gca();
45 xset ('window', 2);
46 plot(frv, hwnvF);
47 plot(frb, hwnbF);
48 xlabel('Digital frequency');
49 ylabel('Magnitude [dB]');
50 title ('Vonhann Fc=0.1313, Minimum N=23, Blackmann Fc
       =0.1278, Minimum N=29');
```

## Scilab code Exa 10.3cd FIR filter Design

```
1 // Design of high pass FIR filter with specifications 2 // fp=4kHZ; fs=2kHZ; Ap=2dB; As=40dB
```

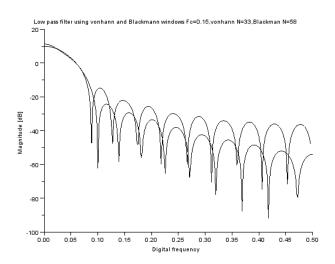


Figure 10.1: FIR lowpass Filter design

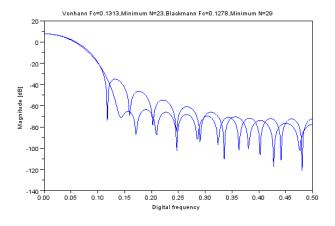


Figure 10.2: FIR lowpass Filter design

```
3 fp=2; fs=4; Ap=2; As=40; S=20;
4 Fp=fp/S;Fs=fs/S;
5 \text{ Ft} = 0.1;
6 \text{ Fc} = 0.15
7 N1=3.47/(Fs-Fp); //hamming
8 \text{ N1} = int(N1) + 1
9 N2=5.71/(Fs-Fp); //blackman
10 N2 = int(N2) + 1
11 [hn1]=eqfir(N1,[0 0.1;0.2 0.5],[0 1],[1 1]);
12 [HF1, fr1] = frmag(hn1, 512);
13 Hf1=20*log10(HF1);
14 [hn2]=eqfir(58,[0 0.1;0.2 0.43],[0 1],[1 1]);
15 [HF2, fr2] = frmag(hn2, 512);
16 Hf2=20*log10(HF2);
17 a=gca();
18 plot2d(fr1, Hf1, rect = [0 -120 0.5 4]);
19 plot2d(fr2(1:length(fr2)-5), Hf2(1:length(fr2)-5),
      rect=[0 -120 0.5 4]);
20 xlabel('Digital Frequency F');
21 ylabel ('Magnitude [dB]');
22 xtitle ('High pass filter using Hamming and Blackmann
       windows LPP Fc=0.35');
23 //Minimum Length Design
24 [hn3]=eqfir(22,[0 0.1;0.2 0.43],[0 1],[1 1]);
25 [HF3,fr3] = frmag(hn3,512);
26 \text{ Hf3}=20*\log 10 \text{ (HF3)};
27 [hn4]=eqfir(29,[0 0.1;0.2 0.5],[0 1],[1 1]);
28 [HF4, fr4] = frmag(hn4, 512);
29 Hf4=20*log10(HF4);
30 xset('window',1);
31 \quad a = gca();
32 plot2d(fr3(1:length(fr3)-5), Hf3(1:length(fr3)-5),
      rect = [0 -120 \ 0.5 \ 4]);
33 plot2d(fr4, Hf4, rect = [0 -120 0.5 4]);
34 xlabel('Digital Frequency F');
35 ylabel('Magnitude [dB]');
36 xtitle('Hamming LPP Fc=0.3293 N=22;Blackmann LPP Fc
      =0.3277 \text{ N}=29');
```

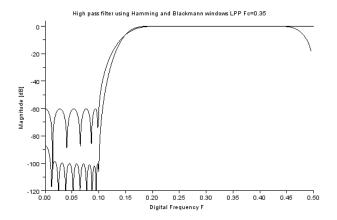


Figure 10.3: FIR filter Design

## Scilab code Exa 10.4a Half Band lowpass FIR filter Design

```
1 // Half band FIR Filter Design
2 //(a)lowpass Half band Filter
3 s=%s;z=%z;
4 fp=8;fs=16;Ap=1;As=50;
5 S=2*(fs+fp);
6 Fp=fp/S;Fs=fs/S;Fc=0.25;
7 delp=(10^(Ap/20)-1)/(10^(Ap/20)+1);
8 dels=10^(-As/20);
9 del=min(delp,dels);
10 As0=-20*log10(del)
11 N1=(As0-7.95)/(14.36*(Fs-Fp))+1;
```

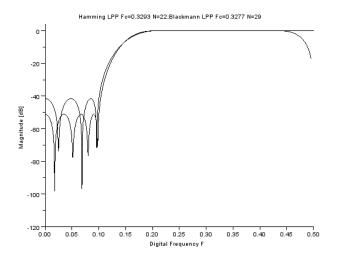


Figure 10.4: FIR filter Design

```
12 N1 = int(N1) + 1;
13 B=0.0351*(As0-8.7)
14 [hn1]=eqfir(19,[0 1/6;1/3 0.5],[1 0],[1 1]);
15 [HLPF1,fr1]=frmag(hn1,512);
16 HLPf1=20*log10(HLPF1);
17 a=gca();
18 plot2d(fr1, HLPf1);
19 xlabel('Digital Frequency');
20 ylabel('Magnitude in dB');
21 xtitle('Kaiser half band LPF:B=1.44;Fc=0.25');
22 [hn2]=eqfir(21,[0 1/6;1/3 0.5],[1 0],[1 1]);
23 [HLPF2, fr2] = frmag(hn2, 512);
24 \text{ HLPf2} = 20 * \frac{\log 10}{\ln \text{HLPF2}};
25 xset('window',1);
26 plot2d(fr2, HLPf2);
27 xlabel('Digital Frequency');
28 ylabel('Magnitude in dB');
29 xtitle('Hamming half-band LPF: N=21; Fc=0.25');
```

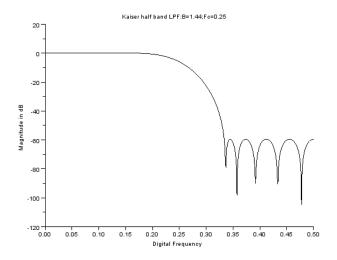


Figure 10.5: Half Band lowpass FIR filter Design

## Scilab code Exa 10.4b Half Band bandstop FIR filter Design

```
1 // Half band FIR Filter Design
2 //(a)band-stop Half band Filter
3 s=%s;z=%z;
4 fp1=1;fs1=2;fp2=4;fs2=3;Ap=1;As=50;
5 S=2*(fs1+fs2);
6 Fp=0.5*(fs2/S-fs1/S);Fs=0.5*(fp2/S-fp1/S);
7 Fc=0.5*(Fp+Fs);Fo=0.25;
8 delp=(10^(Ap/20)-1)/(10^(Ap/20)+1);
9 dels=10^(-As/20);
10 del=min(delp,dels);
11 As0=-20*log10(del)
```

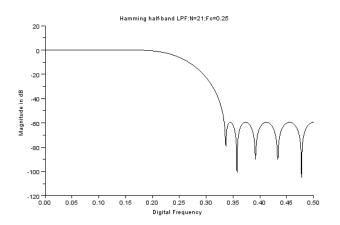


Figure 10.6: Half Band lowpass FIR filter Design

```
12 N1 = (As0 - 7.95) / (14.36*(Fs - Fp)) + 1;
13 N1 = ceil(N1);
14 B=0.0351*(As0-8.7)
   [hn1]=eqfir(31,[0 0.1;0.2 0.3;0.4 0.5],[1 0 1],[1 1
15
      1]);
16 [HBSF1, fr1] = frmag(hn1, 400);
17 HBSf1 = 20 * log10 (HBSF1);
18 a=gca();
19 plot2d(fr1, HBSf1);
20 xlabel('Digital Frequency');
21 ylabel('Magnitude in dB');
22 xtitle('Kaiser half band LPF:B=1.44;Fc=0.25');
23 [hn2]=eqfir(35,[0 0.1;0.2 0.3;0.4 0.5],[1 0 1],[1 1
      1]);
24 [HF2,fr2]=frmag(hn2,200);
25 \text{ HBSf2} = 20 * \log 10 \text{ (HF2)};
26 xset('window',1);
27 plot2d(fr2, HBSf2);
28 xlabel('Digital Frequency');
29 ylabel('Magnitude in dB');
```

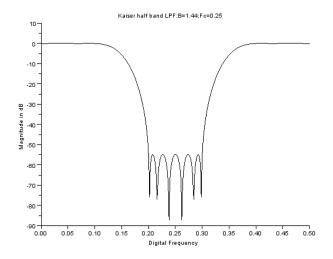


Figure 10.7: Half Band bandstop FIR filter Design

```
30 xtitle('Hamming half-band LPF:N=21;Fc=0.25');
```

## Scilab code Exa 10.5a Design by Frequency Sampling

```
1 //Fir low pass filter design by frequency sampling
2 z=%z;
3 N=10;
4 magHk=[1 1 1 0 0 0 0 0 1 1];
5 k=[0:7 -1 -2];
6 fik=-%pi*k'*(N-1)/N;
7 for i=1:length(fik)
8 H1k(i)=magHk(i)*exp(%i*fik(i));
```

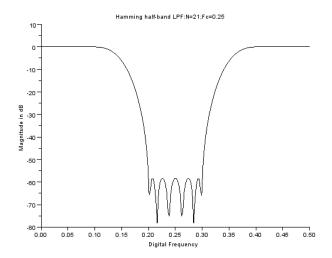


Figure 10.8: Half Band bandstop FIR filter Design

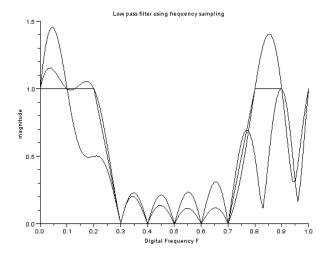


Figure 10.9: Design by Frequency Sampling

```
9 end
10 H1n=(dft(H1k,1));
11 H2k=H1k;
12 H2k(3) = 0.5*\%e^{(-\%i*1.8*\%pi)};
13 H2k(9) = 0.5*\%e^{(\%i*1.8*\%pi)};
14 H2n = (dft(H2k, 1));
15 H1Z=0; H2Z=0;
16 for i=1:length(H1n)
       H1Z=H1Z+H1n(i)*z^{(-i)};
17
18 end
19 for i=1:length(H2n)
       H2Z=H2Z+H2n(i)*z^{(-i)};
20
21 end
22 F=0:0.01:1;
23 F1=0:0.1:0.9;
24 H1F=abs(horner(H1Z,exp(%i*2*%pi*F')));
25 H2F=abs(horner(H2Z,exp(%i*2*%pi*F')));
26 a=gca();
27 plot2d(F1, magHk);
28 plot2d(F, H2F);
29 plot2d(F, H1F);
30 xlabel('Digital Frequency F');
31 ylabel('magnitude');
32 xtitle('Low pass filter using frequency sampling');
```

## Scilab code Exa 10.5b Design by Frequency Sampling

```
1 //Fir high pass filter design by frequency sampling
2 z=%z;
3 N=10;
4 magHk=[0 0 0 1 1 1 1 1 0 0];
5 k=[0:5 -4:-1:-1];
6 fik=(-%pi*k'*(N-1)/N)+(0.5*%pi);
```

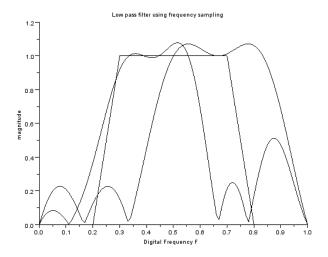


Figure 10.10: Design by Frequency Sampling

```
7 for i=1:length(fik)
       H1k(i) = magHk(i) * exp(%i * fik(i));
8
9 end
10 H1n=(dft(H1k,1));
11 H2k=H1k;
12 H2k(3) = 0.5*\%e^{(-\%i*1.3*\%pi)};
13 H2k(9) = 0.5*\%e^{(\%i*1.3*\%pi)};
14 H2n = (dft(H2k, 1));
15 H1Z=0; H2Z=0;
16 for i=1:length(H1n)
17
       H1Z=H1Z+H1n(i)*z^{(-i)};
18 end
19 for i=1:length(H2n)
       H2Z=H2Z+H2n(i)*z^{(-i)};
20
21 end
22 \quad F = 0:0.01:1;
23 F1=0:0.1:0.9;
24 H1F=abs(horner(H1Z,exp(%i*2*%pi*F')));
25 H2F=abs(horner(H2Z,exp(%i*2*%pi*F')));
26 a=gca();
```

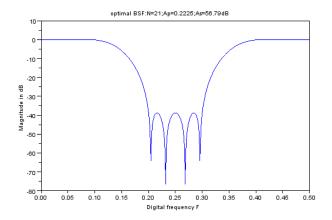


Figure 10.11: Optimal FIR Bandstop Filter Design

```
27 plot2d(F1,magHk);
28 plot2d(F,H2F);
29 plot2d(F,H1F);
30 xlabel('Digital Frequency F');
31 ylabel('magnitude');
32 xtitle('Low pass filter using frequency sampling');
```

## Scilab code Exa 10.6a Optimal FIR Bandstop Filter Design

```
1 //optimal Fir band stop filter design
2 fp1=1;fp2=4;fs1=2;fs2=3;
3 Ap=1;As=50;S=10;
4 Fp1=fp1/S;Fp2=fp2/S;Fs1=fs1/S;Fs2=fs2/S;
5 FT=0.1;FC=0.25
6 //calculation of filter length
7 delp=(10^(Ap/20)-1)/(10^(Ap/20)+1);
```

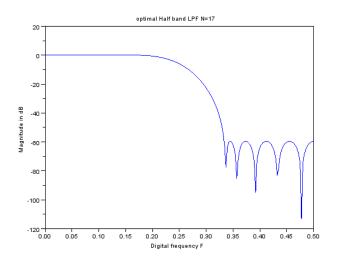


Figure 10.12: Optimal Half Band Filter Design

```
8 dels=10^(-As/20);
9 N=1+((-10*log10(delp*dels)-13)/(14.6*FT))
10 N1=21;
11 [hn]=eqfir(N1,[0 0.1;0.2 0.3;0.4 0.5],[1 0 1],[1 1 1]);
12 [HF,fr]=frmag(hn,512);
13 Hf=20*log10(HF);
14 a=gca();
15 plot(fr,Hf);
16 xlabel('Digital frequency F');
17 ylabel('Magnitude in dB');
18 xtitle('optimal BSF:N=21;Ap=0.2225;As=56.79dB');
```

Scilab code Exa 10.6b Optimal Half Band Filter Design

```
1 //optimal Fir band pass filter design
```

```
2 \text{ fp=8; fs=16;}
3 \text{ Ap=1; As=50; S=48;}
4 Fp=fp/S;Fs=fs/S;
5 FT=0.1; FC=0.25
6 //calculation of filter length
7 delp = (10^{(Ap/20)} - 1)/(10^{(Ap/20)} + 1);
8 \text{ dels}=10^{(-As/20)};
9 del=min(delp,dels);
10 N=1+((-10*log10(del*del)-13)/(14.6*FT));
11 N1=19;
12 [hn]=eqfir(N1,[0 1/6;1/3 0.5],[1 0],[1 1]);
13 [HF, fr] = frmag(hn, 200);
14 Hf = 20 * log 10 (HF);
15 a=gca();
16 plot(fr, Hf);
17 xlabel('Digital frequency F');
18 ylabel ('Magnitude in dB');
19 xtitle('optimal Half band LPF N=17');
```

#### Scilab code Exa 10.7 Multistage Interpolation

```
//The concept of multistage Interpolation
//(a) Single stage interpolator
Sin=4; Sout=48;
fp=1.8;
fs=Sin-fp;
FT=(fs-fp)/Sout;
disp('By using single stage the total filter length is:')
L=4/FT
//(b)Two-stage interpolator
Sin=[4 12];
I=[3 4];//interpolating factors
Sout=[12 48];
fp=[1.8 1.8];
```

```
14 fs=Sin-fp;
15 L1=4*Sout./(fs-fp);
16 L=0;
17 for i=1:length(L1)
18
        L=L+L1(i);
19 end
20 disp('By using 2 stage interpolator filter length is
      :")
21 ceil(L)
22 //(c)3 stage interpolator with I1=2; I2=3; I3=2
23 Sin = [4 \ 8 \ 24];
24 I = [2 \ 3 \ 2];
25 Sout = [8 \ 24 \ 48];
26 fp = [1.8 \ 1.8 \ 1.8];
27 fs = Sin - fp;
28 L2=4*Sout./(fs-fp); L=0;
29 for i = 1: length (L2)
        L=L+L2(i);
30
31 end
32 disp('By using 3 stage interpolator filter length is
      :")
33 ceil(L)
34 / (d) 3 stage interpolator with I1=2; I2=3; I3=2
35 \sin = [4 \ 12 \ 24];
36 \quad I = [3 \quad 2 \quad 2];
37 Sout = [12 \ 24 \ 48];
38 fp = [1.8 \ 1.8 \ 1.8];
39 \text{ fs} = \text{Sin} - \text{fp};
40 L3=4*Sout./(fs-fp); L=0;
41 for i = 1: length (L3)
42
        L=L+L3(i);
43 end
44 disp ('By using 2 stage interpolator filter length is
      :")
45 ceil(L)
```

#### Scilab code Exa 10.8 Design of Interpolating Filters

```
1 //Design of interpolating filters
2 //(a) Design using a single stage interpolator
3 fp=1.8; Sout=48; Sin=4;
4 Ap=0.6; As=50;
5 fs=Sin-fp;
6 //finding ripple parameters
7 delp=(10^{(Ap/20)-1)/(10^{(Ap/20)+1)};
8 \text{ dels}=10^{(-As/20)};
9 N=Sout*(-10*log10(delp*dels)-13)/(14.6*(fs-fp))+1;
10 disp('By using single stage interpolator the filter
      design is: ');
11 ceil(N)
12 // Design using 3-stage interpolator with I1=2; I2=3;
      I3 = 2
13 Ap=0.2;
14 Sin=[4 8 24];
15 Sout = [8 24 48];
16 fp=[1.8 1.8 1.8];
17 fs=Sin-fp;
18 delp = (10^{(Ap/20)-1)/(10^{(Ap/20)+1)};
19 dels=10^{(-As/20)};
20 p=14.6*(fs-fp);
21 N1 = ((-10 * log10 (delp*dels) - 13)./p);
22 \text{ N1} = (\text{Sout.} * \text{N1}) + 1; N = 0;
23 for i=1:length(N1)
24
       N=N+N1(i);
25 end
26 disp('By using single stage interpolator the filter
      design is: ');
27 ceil(N)
```

## Scilab code Exa 10.9 Multistage Decimation

```
1 //The concept of multistage Decimation
2 //(a) Single stage decimator
3 Sin=48; Sout=4;
4 \text{ fp=1.8;}
5 fs=Sout-fp;
6 FT=(fs-fp)/Sin;
7 disp('By using single stage the total filter length
      is: ')
8 L=4/FT
9 //(b)Two-stage decimator
10 Sin=[48 12];
11 D=[4 3]; //decimating factors
12 Sout = [12 4];
13 fp=[1.8 1.8];
14 fs=Sout-fp;
15 L1=4*Sin./(fs-fp);
16 L=0;
17 for i=1:length(L1)
       L=L+L1(i);
18
19 end
20 disp('By using 2 stage decimator filter length is:")
21 ceil(L)
22 //3 stage decimator with D1=2;D2=3;D3=2
23 Sin = [48 \ 24 \ 8];
24 D=[2 3 2];
25 Sout = [24 \ 8 \ 4];
26 fp = [1.8 \ 1.8 \ 1.8];
27 fs = Sout - fp;
28 L2=4*Sin./(fs-fp); L=0;
29 for i = 1: length (L2)
30
       L=L+L2(i);
31 end
```

## Scilab code Exa 10.10 Maximally Flat FIR filter Design

```
//Maximally flat FIR filter design
Fp=0.2;
Fs=0.4;
Fc=0.3;
Ft=0.2;
No=1+0.5/Ft^2;
No=ceil(N0);
alpha=(cos(%pi*Fc))^2;
k=5; Mmin=14;
L=Mmin-k;
N=2*Mmin-1;
disp(N,'Hence with this length we can get maximally flat FIR filter with no ripples in passband');
```

# **Appendix**

## Scilab code AP 1 Alias Frequency

```
1 function[F]=aliasfrequency(f,s,s1)
2 if (s>2*f) then
       disp("alias has not occured")
3
4 else
       disp("alias has occured")
5
6 end
7 \text{ F=f/s};
8 for i=1:100
       if (abs(F) > 0.5)
           F=F-i;
10
11
       end
12 end
13 fa=F*s1;
14 disp(fa, "frequency of reconstructed signal is")
15 endfunction
```