EL 7133 (DSP II) Spring 2016

Homework Assignment - Week 03

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Question Part B

The problem: For d = 1, 3, 5, find Q(z) such H(z) above is a Type I halfband filter. For d = 1 the problem is trivial. For d = 3 you can solve the linear equations problem by hand. For d = 5 you can use MATLAB to solve the system of linear equations.

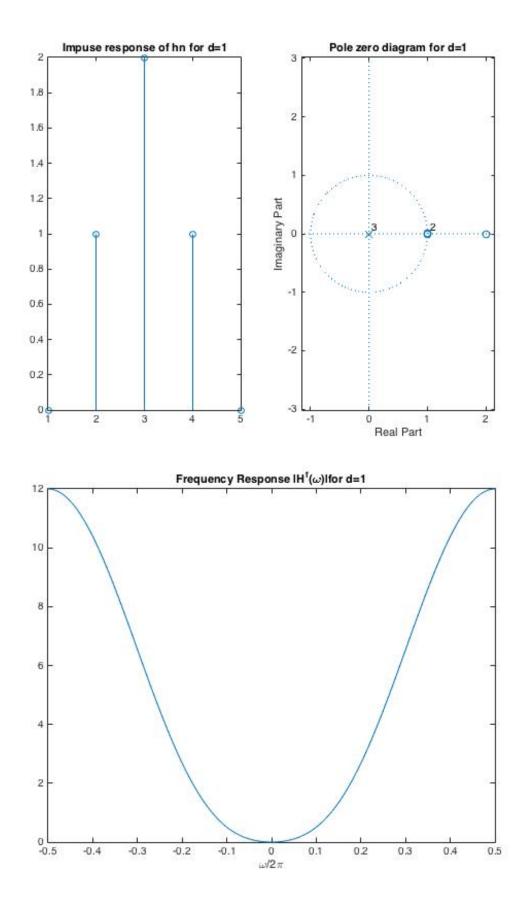
In each case, check that h(n) really is a halfband filter, make a stem plot of h(n), and include the zero diagram and frequency response $|Hf(\omega)|$. 1

Solution

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Program
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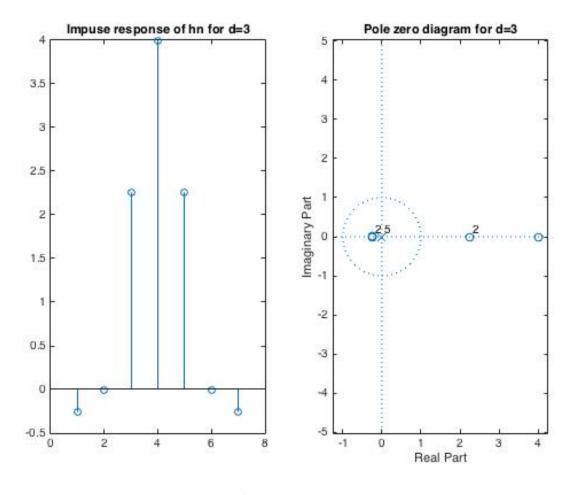
```
For d=1;
clc; clear all; close all;
% for d=1
d=1;
q=[0.5 0]
p=[1 1];
p1=[1 1];
for i=1:d
    p=conv(p,p1);
end
hn=conv(q,p)
b=poly(hn);
a=1;
% Stem plot
figure, subplot(1,2,1), stem(hn);
title(['Impuse response of hn for d=' num2str(d)]);
% Pole zero diagram
subplot(1,2,2), zplane(b,a)
title(['Pole zero diagram for d=' num2str(d)]);
% Frequency Response
j=sqrt(-1);
om=linspace(-pi,pi,200);
Hf=polyval(b,exp(j*om))./polyval(a,exp(j*om));
figure, plot(om/(2*pi), abs(Hf))
title(['Frequency Response | H^f(\omega)| for d=',num2str(d)]);
xlabel('\omega/2\pi');
```

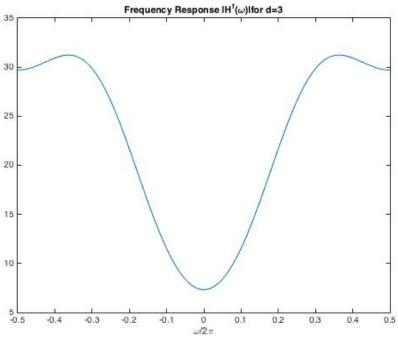
Result:



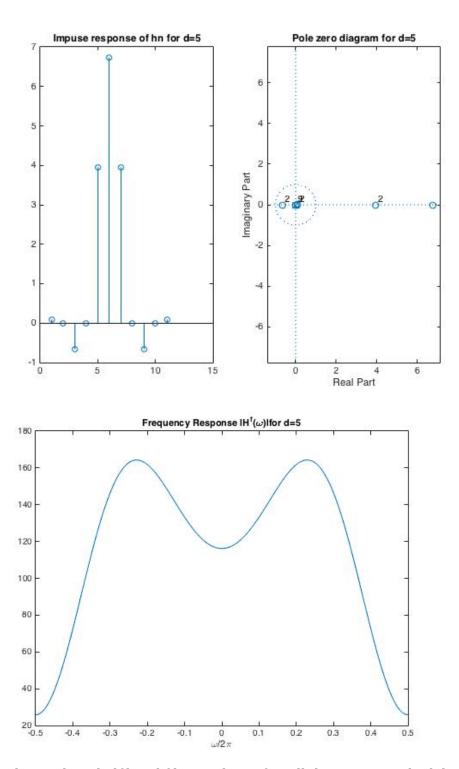
H(n) is half band

q=[-1/4 1 -1/4]





q=[3/38 -9/19 1 -9/19 3/38]

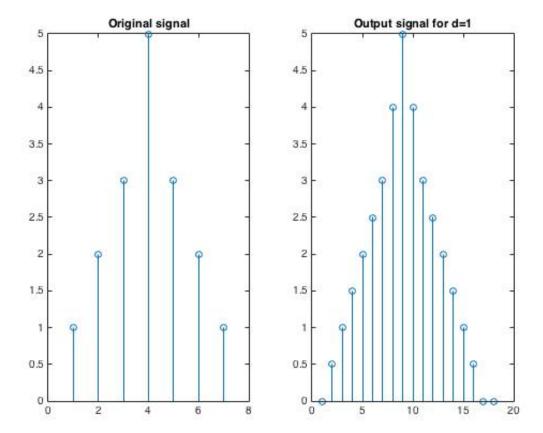


h(n) obtained is a half band filter and satisfies all the constraint laid down in the design of filter.

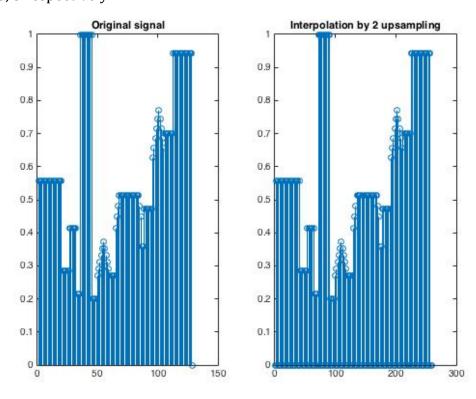
C) Perform interpolation (by factor 2) using the half-band filters from (B). Verify that the values of the input signal are not changed by the process. Use the signal provided on the web: bumps.txt, pwsmooth.txt, skyline.txt. Comment on your results.

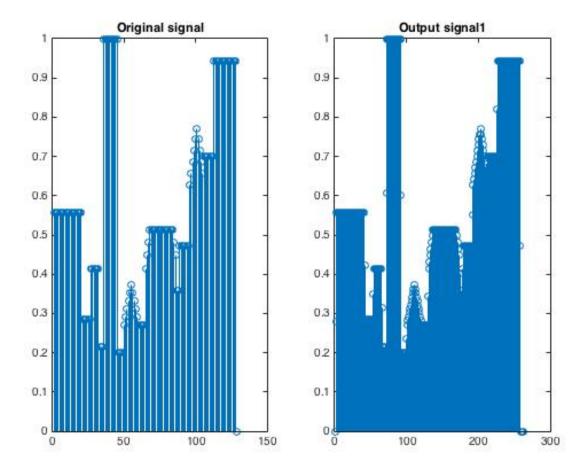
```
clc; clear all; close all;
% for d=1
d=1;
q=[0.5 0]
q=[-1/4 \ 1 \ 1/4];
% q=[3/38 -9/19 1 -9/19 3/38]
p=[1 1];
p1=[1 1];
for i=1:d
    p=conv(p,p1);
end
hn=conv(q,p)
% Input signal
% x=[1 2 3 5 3 2 1];
x=dlmread('skyline.txt');
% Performing interpolation by 2
l=length(x);
if mod(length(x), 2) == 0
  x(1+1)=0;
end
l=length(x)-1;
j=1+1;
for i=1/2:-1:1
    x1(j)=0;
    x1(j-1)=x(i);
    j=j-2;
end
j=1+2;
for i=1/2+1:1+1
    x1(j)=x(i);
    x1(j+1)=0;
    j=j+2;
end
% Interpolation figure
figure,
subplot(1,2,1),stem(x);title('Original signal');
subplot(1,2,2),stem(x1);title('Interpolation by 2 upsampling')
% Output
yn=conv(x1,hn);
figure,
subplot(1,2,1),stem(x);title('Original signal');
subplot(1,2,2),stem(yn);title(['Output signal for d='num2str(d)]);
```

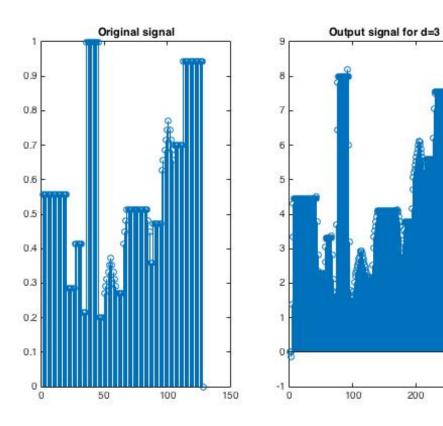
First I took my own signal as input Here, is the result

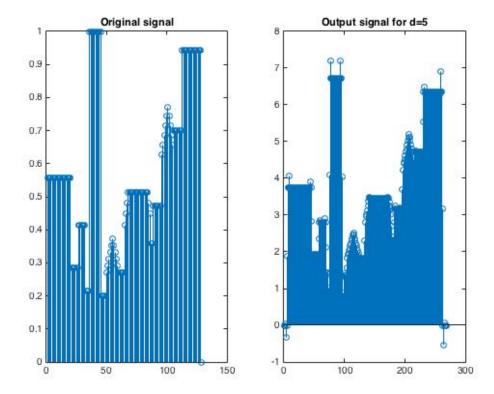


For skyline input, For d=1,3, 5 respectively









Here, first we performed the upsampling by a factor of 2 and then did the interpolation by the designed filter for d=1, 3, 5 The result obtained is perfect for d=1. However, The result obtained is modified by some magnitude uniformly for d=3 and 5 after convolution which is not a desired property.