

DSP lab3 Report

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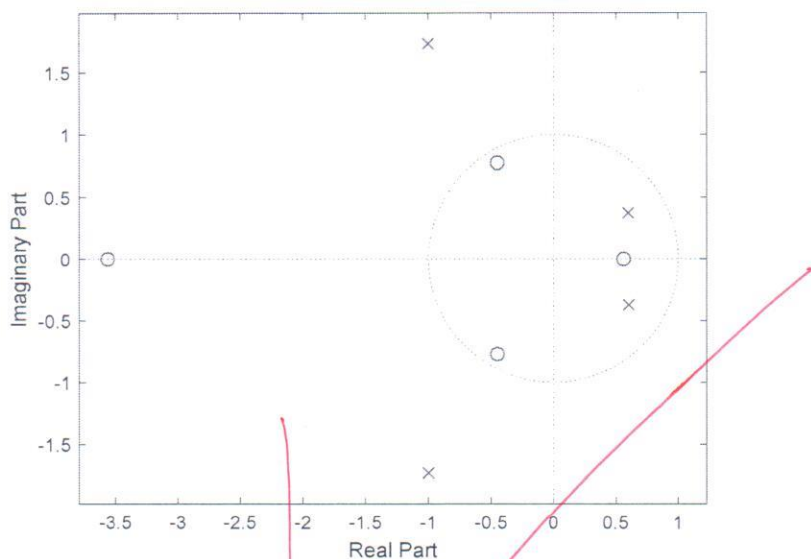
1. Using MATLAB determine the second-order factored form of the following z-transforms. And show their pole-zero plots. Then determine all possible ROCs of above z-transform.

$$G(z) = \frac{4z^4 + 15.6z^3 + 6z^2 + 2.4z - 6.4}{3z^4 + 2.4z^3 + 6.3z^2 - 11.4z + 6}$$

(1) Code:

```
num=[4 15.6 6 2.4 -6.4];%the numerator coefficients
den=[3 2.4 6.3 -11.4 6];%the denominator coefficients
[sos,G]=tf2sos(num,den);
[z,p,k]=tf2zp(num,den);%calculating the zeros and poles
zplane(z,p);
```

(2) Result



The coordination of poles are

$-0.9999999999999999 + 1.73205080756888i$

$-0.9999999999999999 - 1.73205080756888i$

$0.6000000000000000 + 0.374165738677394i$

$0.6000000000000000 - 0.374165738677394i$

The magnitudes are

2.0000

2.0000

0.7071

0.7071

ROC are:

$|z| < 0.7071$; $0.7071 < |z| < 2$; $|z| > 2$

2. Writing a MATLAB program to determine the rational form of a z-transform whose zero are at $\xi_1 = 1.2$, $\xi_2 = 2.3 - j0.5$, $\xi_3 = -0.4 + j0.2$, $\xi_4 = -0.4 - j0.2$, $\xi_5 = 2.3 + j0.5$; the poles are at $\lambda_1 = 0.5$, $\lambda_2 = -0.75 + j0.2$, $\lambda_3 = 0.6 + j0.7$, $\lambda_4 = 0.6 - j0.7$, $\lambda_5 = -0.75 - j0.2$; and the gain constant k is 2.1.

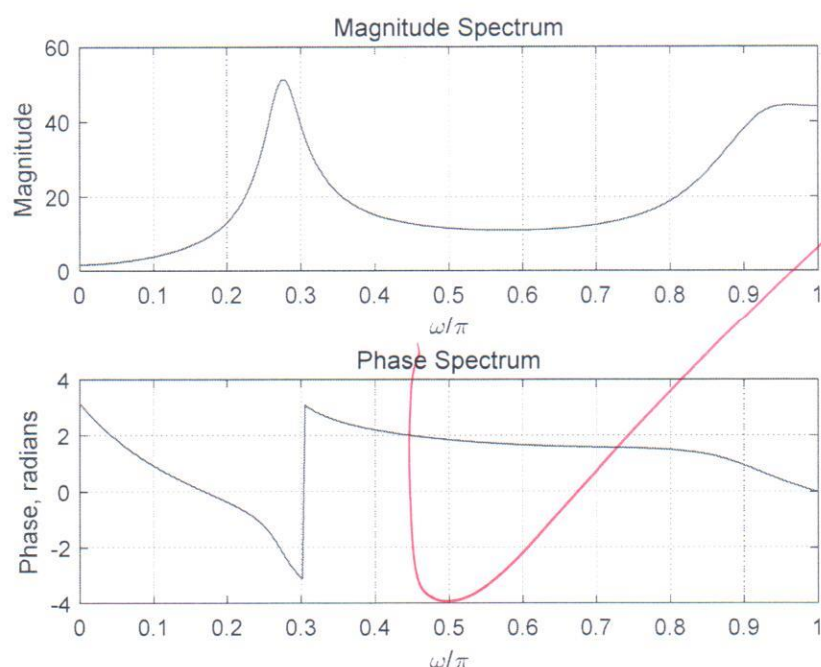
(1) Code

```
%the coordination of zeros and poles
z=input('input the zeros=');
p=input('input the ploes=');
g=input('input the gain constant=');
[num,den]=zp2tf(z',p',g);%calculating the corresponding coefficients
w = 0:pi/(255):pi;%the frequency response range
h=freqz(num,den,w);
subplot(2,1,1)
plot(w/pi,abs(h));grid
title('Magnitude Spectrum')
xlabel('\omega/\pi'); ylabel('Magnitude')
subplot(2,1,2)
plot(w/pi,angle(h));grid
title('Phase Spectrum')
xlabel('\omega/\pi'); ylabel('Phase, radians')
```

Entering the following commands

```
input the zeros=[1.2 2.3-0.5*j -0.4+0.2*j -0.4-0.2*j 2.3+0.5*j]
input the ploes=[0.5 -0.75+0.2*j 0.6+0.7*j 0.6-0.7*j -0.75-0.2*j]
input the gain constant=2.1
```

(2) Result:the frequency response of the system



3. Using the function `fir1` and window of Kaiser, design a linear-phase FIR lowpass filter meeting the following specifications: passband edge frequency = 2 kHz, stopband edge frequency=2.5 kHz, passband ripple $\delta_p=0.005$, stopband ripple $\delta_s=0.005$, and sampling rate of 10 kHz. Plot its gain and phase responses and check if it meets the specifications?

(1) Code

```
[n,wn,beta,typ]=kaiserord([2000 2500],[1 0],[0.005 0.005],10000);
%entering the parameters of the systems
b=fir1(n,wn,kaiser(n+1,beta),'noscale');
```

```
[h,omega]=freqz(b,1,256);%the frequency response
subplot(2,1,1)
plot(omega/pi,20*log10(abs(h)));
xlabel('\omega/\pi'); ylabel('Gain, dB');
subplot(2,1,2)
plot(omega/pi,angle(h));grid
title('Phase Spectrum')
xlabel('\omega/\pi'); ylabel('Phase, radians')
```

(2) Result: the frequency response of the system

