



## TTT4120 Digital Signal Processing Problem Set 4

### Problem 1 (2 points)

Given a filter with transfer function

$$H(z) = \frac{1}{1 - az^{-1}}$$

- (a) Draw the pole-zero plot for the filter given  $a = 0.9$  and  $a = -0.9$ .  
Determine the filter type for two filters? Explain using the pole-zero plot.
- (b) Verify the results in 1(a) with *pezdemo*. The demo can be downloaded from the course home page.

### Problem 2 (2 points)

Consider a causal digital filter with transfer function

$$H(z) = \frac{1}{(1 - \frac{1}{2}z^{-1})(1 + \frac{1}{2}z^{-1})}$$

- (a) Find the transfer function of the inverse filter of  $H(z)$
- (b) Is the inverse filter stable? Justify the answer.
- (c) Is the inverse filter a minimum-phase filter?
- (d) Does the inverse filter have a linear phase characteristics? Justify your answer.

### Problem 3 (2 points)

In the recording/mastering of sound signals or during playback, it is often desired to alter the characteristics of the sound at different frequencies. For

example, we may wish to highlight the lower/middle frequencies, while we may wish to reduce the presence of high frequencies.

This can be done by using so-called 'shelving' filters. Figure 1 shows a low-frequency shelving filter implementation. The filter  $A(z)$  is :

$$A(z) = \frac{\alpha - z^{-1}}{1 - \alpha z^{-1}}$$

The parameters  $\alpha$  and  $K$  are used to *tune* the filter.

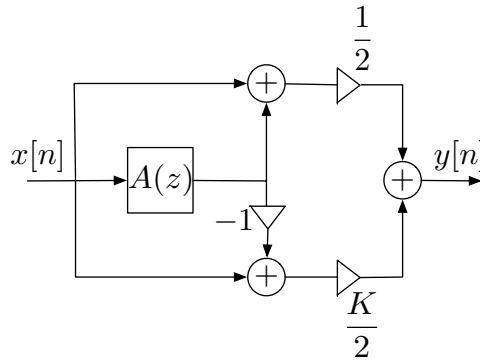


Figure 1: Low-frequency shelving filter

- (a) What type of filter is  $A(z)$ , (Highpass, Lowpass, Bandpass, Bandstop or Allpass)? Justify your answer.
- (b) The filter in Figure 1 consists of a sum of two branches (upper and lower).
  - Use Matlab function `freqz` to plot the magnitude responses of the two branches given  $\alpha = 0.9$  and  $K = 1$ .
  - What types of filters do the upper and lower branches represent?
- (c) The Matlab-script `LFshelving.m` implements the entire filter in Figure 1 and plots its magnitude response. Furthermore, it uses the filter to modify the music file `pluto.wav` and plays both the original and modified music file.
  - Let  $K = 3$ . Plot the magnitude response of the filter and listen to the original and modified music file when  $\alpha$  is equal to 0.5, 0.7 and 0.9, respectively.
  - Let  $\alpha = 0.7$ . Plot the magnitude response of the filter and listen to the original and modified music file when  $K$  is equal to 0.5, 1 and 4, respectively.
  - What do the parameters  $K$  and  $\alpha$  control?

#### Problem 4 (4 points)

Given a sequence  $d(n)$  as:

$$d(n) = A_x \cos(2\pi f_x n) + A_y \cos(2\pi f_y n), \quad 0 \leq n \leq L - 1$$

where  $A_x = A_y = 0.25$ ,  $f_x = 0.04$ ,  $f_y = 0.10$  and  $L = 500$ .

The sequence  $d(n)$  is contaminated with additive noise  $e(n)$ , that is, the observed signal is

$$g(n) = d(n) + e(n).$$

- (a) Use MATLAB to generate and plot sequences  $d(n)$  and  $g(n)$  and their magnitude spectra,  $|D(f)|$  and  $|G(f)|$ . (Use FFT length  $N=2048$ )  
(A segment of the noise  $e(n)$  of length  $L$  can be generated by MATLAB command `randn(1,L)`)  
Compare the plots before and after adding the noise.
- (b) To isolate the two sinusoids from the noisy signal  $g(n)$  we want to design two digital resonators with transfer functions  $H_x(z)$  and  $H_y(z)$ . The resonators should have zeros at  $z = 1$  and  $z = -1$ . Use common sense to figure out how close to the unit circle the poles should be.
- Write the expressions for  $H_x(z)$  and  $H_y(z)$ .
  - Read about the Matlab functions `poly`, `roots`, `zplane` and `freqz`.
  - Use `zplane` to plot the zeros and poles of the resonators.
  - Use `freqz` to plot  $|H_x(f)|$  and  $|H_y(f)|$ .
- (c) Use the two filters designed in 2b) to filter the noise contaminated signal  $g(n)$  (use the Matlab function `filter`)  
Plot the outputs from the filters  $q_x(n)$  and  $q_y(n)$  as well as their amplitude spectra  $|Q_x(f)|$  and  $|Q_y(f)|$ .  
Are the resulting plots what you expected?
- (d) We wish to combine the two digital resonators in order to isolate both sinusoids.
- Plot the magnitude response of the resulting system.
  - Find its zeros and poles. (Hint. You can use the MATLAB functions `poly` and `roots`)
  - Use `zplane` to plot the zeros and poles, and discuss their placement.
  - Plot the output from the combined filter, and the its magnitude spectra.
  - Compare the plots with the plots of  $d(n)$  and  $g(n)$  and their magnitude spectra.