

# Tik-61.246 Digital Signal Processing and Filtering

2. Mid Term Exam 13.12.1999 at 9-12. Halls A, C, D.

1. Are the following statements right or wrong? (A right answer gives +1 points, a wrong answer -1 points, no answer 0 points. The minimum is still 0 points and the maximum 3 points.)
    - a) When converting an analog IIR type low pass filter to a digital filter using the bilinear transform, the order of the filter almost always increases.
    - b) With the impulse-invariant method, there is a one-to-one mapping from the frequency axis of the  $s$ -plane to the unit circle in the  $z$ -plane.
    - c) If we use a Hamming window with length 25 to design a linear phase FIR filter, the group delay is always 12.
    - d) By scaling a filter  $H(z)$  with  $\max\{|KH(z)|\} = 1$ , we can prevent overflowing and improve the signal-to-noise ratio.
- (3p)

2. Suppose you want to design a first-order stable and causal high-pass filter having the following transfer function:

$$H(z) = K \frac{1 + bz^{-1}}{1 + az^{-1}},$$

where the coefficients  $a$ ,  $b$ , and  $K$  are real. The specifications of the filter are as follows:

- 1) the amplitude response is zero at zero frequency ( $|H(e^{j0})| = 0$ ),
- 2) the amplitude response is one at  $\frac{1}{2}$  of the sampling frequency ( $|H(e^{j\pi})| = 1$ ), and
- 3) the attenuation at  $\frac{1}{4}$  of the sampling frequency is 20 dB ( $|H(e^{j\frac{\pi}{2}})| = 0.1$ ).

Determine the coefficients of the filter, draw the zero-pole diagram, and sketch the amplitude response.

(6p)

3. Consider two finite impulse response (FIR) systems having the following impulse responses:

$$\begin{aligned} h_1[n] &= \delta[n] + 2\delta[n - 2] + \delta[n - 4] \\ h_2[n] &= \delta[n] - \delta[n - 4] \end{aligned}$$

- a) Determine the impulse response  $h_c[n]$  and the transfer function  $H_c(z)$  of the cascade connection of the above systems (scaling to unity is not required). Is the phase response linear? What is the value of the phase response at frequency  $\omega = 0$ ? Explain the results or calculate them!
- b) Determine the impulse response  $h_p[n]$  and the transfer function  $H_p(z)$  of the parallel connection of the above systems (scaling to unity is not required). Is the phase response linear? What is the value of the phase response at frequency  $\omega = 0$ ? Explain the results or calculate them!
- c) What is the unit step response of the cascade connection, i.e. the response of the system to the unit step sequence  $\mu[n] = 1, n \geq 0, \mu[n] = 0, n < 0$ . What is the unit step response like when  $n$  is large? Why?

(6p)

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4. In the figure below, there is a flow chart illustrating a second-order digital IIR type filter. Draw all the possible locations of the poles ( $p_1, p_2$ ) of the filter when the real coefficients  $a_1$  and  $a_2$  are quantized to three bits using sign-magnitude truncation. With the used precision we can thus represent the coefficient values  $[-\frac{3}{4}, -\frac{1}{2}, -\frac{1}{4}, 0, \frac{1}{4}, \frac{1}{2}, \frac{3}{4}]$ . Note that due to the real coefficients, the poles are complex conjugates:  $p_1 = r e^{j\theta}, p_2 = r e^{-j\theta}$ .
- How does the quantization of the coefficients affect the realization of the filter at different values of the frequency  $\omega$ ? Consider situations when a narrow-banded low-pass filter or a narrow-banded band-pass filter (with a passband at  $\frac{1}{2}$  of the sampling frequency) is desired.

(6p)

