

**Partial exam EE2S11 Signals and Systems**  
**Part 2: 3 February 2017, 13:30–15:30**

Closed book; two sides of handwritten notes permitted

This exam consists of five questions (40 points)

**Question 1 (14 points)**

- a) Given the signals  $x[n] = u[n + 2] - u[n - 2]$ ,  $h[n] = [\dots, 0, \boxed{1}, -2, 0, 0, \dots]$ .  
 Determine  $y[n] = x[n] * h[n]$  using the convolution sum (in time-domain).
- b) Given  $x[n] = u[n] - (\frac{1}{2})^n u[n - 4]$ . Determine  $X(z)$  and also specify the ROC.
- c) Given  $X(z) = \frac{4z^{-1}}{(1 - z^{-1})(1 + 0.25z^{-1})}$ , ROC =  $\{0.25 < |z| < 1\}$ .  
 Determine  $x[n]$  using the inverse  $z$ -transform.
- d) Given  $x[n] = a^n u[n]$  with  $|a| < 1$ . Determine  $y[n] = x[n] * x[-n]$ . (Use the  $z$ -transform.)
- e) Determine, if it exists, the frequency response  $H(e^{j\omega})$  for the system defined by the difference equation

$$y[n] = 1.6y[n - 1] - 0.64y[n - 2] + x[n] - x[n - 2]$$

- f) Given an LTI system with transfer function  $H(z) = 1 - 2z^{-1}$ .  
 Determine a (bounded) input signal  $x[n]$  for which the output signal is equal to  $y[n] = \delta[n] + \frac{1}{2}\delta[n - 1]$ .

**Solution**

a)  $y[n] = \sum_k h[k]x[n - k]$

$h[0]x[n] :$	$\dots$	0	1	1	$\boxed{1}$	1	1	0	$\dots$
$h[1]x[n - 1] :$	$\dots$	0	-2	$\boxed{-2}$	-2	-2	-2	0	$\dots$
$y[n] :$	$\dots$	0	1	-1	$\boxed{-1}$	-1	-1	-2	$0 \dots$

b)

$$x[n] = u[n] - (\frac{1}{2})^4 (\frac{1}{2})^{n-4} u[n - 4]$$

$$X(z) = \frac{1}{1 - z^{-1}} - \frac{(\frac{1}{2})^4 z^{-4}}{1 - \frac{1}{2}z^{-1}}$$

ROC:  $\{|z| > 1\}$

c)

$$X(z) = \frac{4z^{-1}}{(1 - z^{-1})(1 + 0.25z^{-1})} = \frac{16/5}{1 - z^{-1}} - \frac{16/5}{1 + 0.25z^{-1}}$$

Check the ROC: the first term is anticausal. Hence we write

$$X(z) = \frac{-16/5 z}{1 - z} - \frac{16/5}{1 + 0.25z^{-1}}$$

$$x[n] = -\frac{16}{5}u[-n - 1] - \frac{16}{5}(-0.25)^n u[n]$$

d)

$$x[n] = a^n u[n] \quad \rightarrow \quad X(z) = \frac{1}{1 - az^{-1}} \quad \text{ROC: } \{|z| > a\}$$

$$x[-n] = a^{-n} u[-n] \quad \rightarrow \quad X(z^{-1}) = \frac{1}{1 - az} \quad \text{ROC: } \{|z| < 1/a\}$$

$$Y(z) = \frac{1}{1 - az^{-1}} \cdot \frac{1}{1 - az} \quad \text{ROC: } \{a < |z| < 1/a\}$$

We now would like to apply a partial fraction expansion, therefore we first write the function using polynomials in  $z$  or  $z^{-1}$  (but not both). Hence:

$$Y(z) = \frac{z}{(z-a)(1-az)} = \frac{A}{z-a} + \frac{B}{1-az} = \dots = \frac{1}{1-a^2} \left( \frac{a}{z-a} + \frac{1}{1-az} \right)$$

Check the ROC to determine which part is causal and which part is anti-causal. The first term is causal and therefore we write it in terms of  $z^{-1}$ . This results in:

$$Y(z) = \frac{1}{1-a^2} \left( \frac{az^{-1}}{1-az^{-1}} + \frac{1}{1-az} \right) = \frac{1}{1-a^2} \left( \frac{1}{1-az^{-1}} - 1 + \frac{1}{1-az} \right)$$

$$y[n] = \frac{1}{1-a^2} (a^n u[n] - 1 + a^{-n} u[-n]) = \frac{a^{|n|}}{1-a^2}$$

e)

$$H(z) = \frac{1 - z^{-2}}{1 - 1.6z^{-1} + 0.64z^{-2}} = \frac{1 - z^{-2}}{(1 - 0.8z^{-1})^2}$$

The poles are  $z_{1,2} = 0.8$  (double), the unit circle is in the ROC and the Fourier transform exists. This results in

$$H(e^{j\omega}) = \frac{1 - e^{-2j\omega}}{(1 - 0.8e^{-j\omega})^2}$$

f)

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

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

Because we would like a bounded  $x[n]$ , we rewrite this as

$$X(z) = \frac{z(1 + \frac{1}{2}z^{-1})}{z - 2} = -\frac{1}{2} \frac{\frac{1}{2} + z}{1 - \frac{1}{2}z} = -\frac{1}{2} \left( \frac{1}{2} + \frac{5}{4} \frac{z}{1 - \frac{1}{2}z} \right)$$

$$x[n] = -\frac{1}{2} \left( \frac{1}{2} \delta[n] + \frac{5}{4} \left( \frac{1}{2} \right)^{-n-1} u[-n-1] \right) = -\frac{1}{4} \delta[n] + \frac{5}{8} \left( \frac{1}{2} \right)^{-n-1} u[-n-1]$$

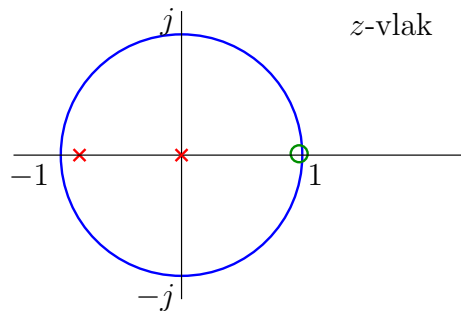
## Question 2 (6 points)

The transfer function of a causal LTI system is given by  $H(z) = \frac{z-1}{z(z+0.9)}$

- Determine all poles and zeros of the system and make a drawing in the complex  $z$ -plane.
- Specify the ROC.
- Is the system BIBO stable? (Why?)
- Draw, based on the poles and zeros of  $H(z)$ , the amplitude response. Is this a low-pass, high-pass or other kind of filter?

**Solution**

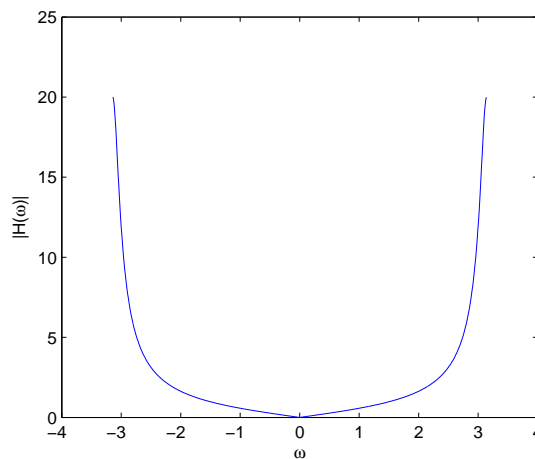
a) Poles:  $z = 0, z = -0.9$ . Zeros:  $z = 1, z = \infty$ .



b) Causal results in ROC:  $|z| > 0.9$ .

c) Unit circle in ROC: BIBO stable.

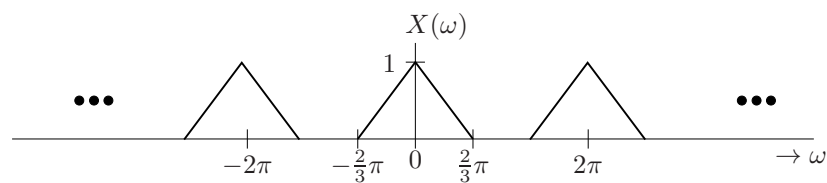
d)



High-pass. Zero at  $z = 1$  results in  $H(e^{j\omega}) = 0$  for  $\omega = 0$ . The pole at  $z = -0.9$  results in a peak for  $\omega = \pm\pi$ . Compute:  $H(e^{j\pi}) = H(1) = 20$ .

**Question 3 (7 points)**

An analog signal  $x_a(t)$  with Fourier transform  $X_a(\Omega)$  is band-limited at 10 kHz. The signal is sampled without aliasing at a sampling frequency  $F_s$ , resulting in the discrete-time signal  $x[n]$ . The spectrum  $X(\omega)$  of  $x[n]$  is shown below:



a) What is the relation between  $\Omega$  and  $\omega$ ?

b) Which sampling frequency was used?

- c) What is the smallest frequency at which we can sample  $x(t)$  without aliasing?  
 For this case, draw the resulting spectrum (also clearly mark the frequencies).
- d)  $x_a(t)$  is reconstructed from  $x[n]$  by means of an ideal D/A convertor and a low-pass filter.  
 Specify the pass-band and stop-band frequencies of the filter.

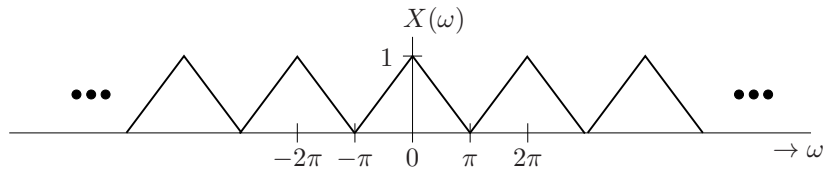
**Solution**

- a)  $\omega = 2\pi \leftrightarrow \Omega = 2\pi F_s$ . This results in  $\Omega = \omega F_s$  i.e.,  $\omega = \Omega T_s$ .
- b)  $\omega = \frac{2}{3}\pi$  results in  $\Omega = \frac{2}{3}\pi F_s$ . At  $F = 10$  kHz we find

$$F = \frac{\Omega}{2\pi} = \frac{\frac{2}{3}\pi F_s}{2\pi} = \frac{1}{3}F_s = 10\text{kHz}$$

hence  $F_s = 30$  kHz.

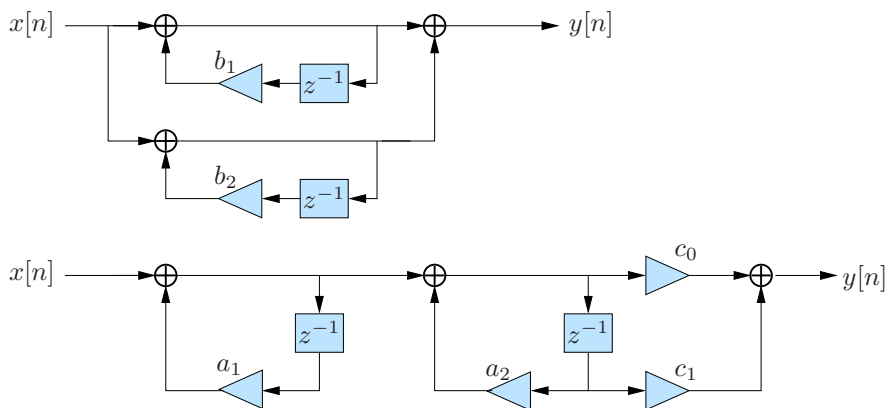
- c)  $F_s = 20$  kHz.



- d) After D/A conversion, the signal is analog. In the frequency spectrum, the frequency  $\omega_p = \frac{2}{3}\pi$  corresponds to  $\Omega_p = 10$  kHz, and the frequency  $\omega_s = \frac{4}{3}\pi$  corresponds to  $\Omega_s = 20$  kHz. The low-pass filter (in the analog domain!) thus has a pass-band running until 10 kHz and a stop-band starting at 20 kHz.

**Question 4 (6 points)**

Given the realizations:



- a) Determine  $a_1, a_2$  and  $c_0, c_1$  in terms of  $b_1, b_2$  such that both systems are equivalent.
- b) Is this a minimal realization?
- c) Draw the “direct form no. II” realization and also specify the coefficients.

**Solution**

a) First realization:

$$H(z) = \frac{1}{1 - b_1 z^{-1}} + \frac{1}{1 - b_2 z^{-1}} = \frac{2 - (b_1 + b_2)z^{-1}}{(1 - b_1 z^{-1})(1 - b_2 z^{-1})}$$

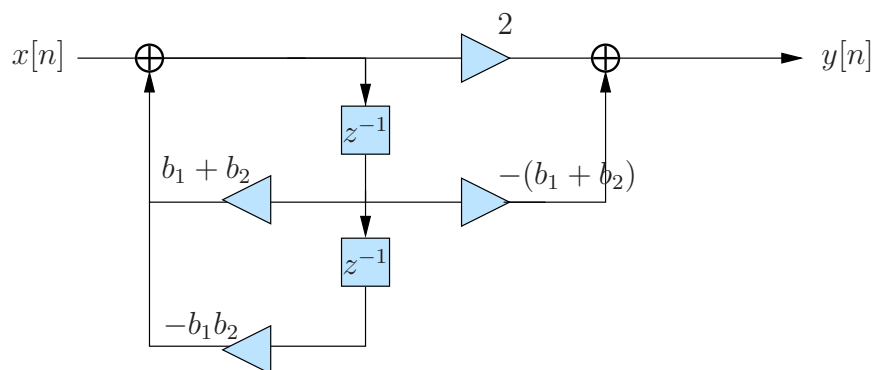
Second realization:

$$H(z) = \frac{1}{1 - a_1 z^{-1}} \cdot \frac{c_0 + c_1 z^{-1}}{1 - a_2 z^{-1}} = \frac{c_0 + c_1 z^{-1}}{(1 - a_1 z^{-1})(1 - a_2 z^{-1})}$$

From this it follows that  $a_1 = b_1$ ,  $a_2 = b_2$ ,  $c_0 = 2$ ,  $c_1 = -(b_1 + b_2)$ .

b) Both are minimal because the number of delays in the realization is equal to the filter order of  $H(z)$ .

c)



**Question 5 (7 points)**

A “template” third-order Butterworth filter has the transfer function

$$H(s) = \frac{1}{s^3 + 2s^2 + 2s + 1}$$

The corresponding frequency response is  $|H(j\Omega)|^2 = \frac{1}{1 + \Omega^6}$ .

- Which frequency transform should we apply to the template to construct a low-pass Butterworth filter with a 3dB cut-off frequency of  $\Omega_c$ ?
- What is the corresponding transfer function  $G(s)$ ?

We now design an analog 3rd order low-pass Butterworth filter with a pass-band frequency of 3 rad/s, a stop-band frequency of 6 rad/s and a maximal damping in the pass-band of 0.5 dB.

- Give a suitable expression for the frequency response (squared-amplitude) of this filter and determine its parameters.
- For this filter, what is the minimal damping in the stop-band ?
- Which transform should be applied to  $|H(j\Omega)|^2$  to obtain this filter?  
Determine the corresponding transfer function.

### Solution

a) Substitute  $\Omega \rightarrow \frac{\Omega}{\Omega_c}$ .

b) Substitute  $s \rightarrow \frac{s}{\Omega_c}$ , this results in

$$G(s) = \frac{1}{\left(\frac{s}{\Omega_c}\right)^3 + 2\left(\frac{s}{\Omega_c}\right)^2 + 2\left(\frac{s}{\Omega_c}\right) + 1}$$

c) The general expression is

$$|H(j\Omega)|^2 = \frac{1}{1 + \epsilon^2 \left(\frac{\Omega}{\Omega_p}\right)^6}$$

For  $\Omega = \Omega_p = 3$  we obtain

$$\frac{1}{1 + \epsilon^2} = 10^{-0.5/10} \Rightarrow \epsilon = 0.3493$$

d) For  $\Omega = \Omega_s = 6$  we obtain

$$\frac{1}{1 + \epsilon^2 \left(\frac{6}{3}\right)^6} = 0.1135 = -9.45 \text{ dB}$$

e) First, we determine  $\Omega_c$ :

$$\left(\frac{\Omega}{\Omega_c}\right)^6 = \epsilon^2 \left(\frac{\Omega}{\Omega_p}\right)^6 \Rightarrow \Omega_c = \frac{\Omega_p}{\epsilon^{1/3}} = 4.26 \text{ rad/s}$$

The transfer function of the requested Butterworth filter is:

$$H(s) = \frac{1}{\left(\frac{s}{4.26}\right)^3 + 2\left(\frac{s}{4.26}\right)^2 + 2\left(\frac{s}{4.26}\right) + 1}$$