# **EE2S11 SIGNALS AND SYSTEMS**

Final exam, 29 January 2021, 13:30–15:50 Block 1 (13:30-14:30)

Open book, strictly timed take-home exam. (Electronic) copies of the book and the course slides allowed. No other tools except a basic pocket calculator permitted. Upload answers during 14:25–14:40

This block consists of three questions (20 points); more than usual, and this will be taken into account during grading. Answer in Dutch or English. Make clear in your answer how you reach the final result; the road to the answer is very important. Write your name and student number on each sheet.

## Question 1 (9 points)

- (a) Let  $x[n] = [\cdots, 0, \boxed{1}, 3, 2, 0, \cdots]$ , where the 'box' denotes the value for n = 0. Determine the convolution y[n] = x[n] \* x[-n].
- (b) Let x[n] = 2<sup>-n-2</sup>u[-n-2]. Determine the z-transform, also specify the ROC. Hint: you could first make a plot of x[n].
- (c) Let

$$H(z) = \frac{1 - \frac{3}{2}z^{-2}}{(1 - \frac{1}{2}z^{-1})(1 - 3z^{-1})}.$$

Draw a pole-zero plot, and determine h[n] for (c1) ROC:  $|z| < \frac{1}{2}$ ; (c2) ROC:  $\frac{1}{2} < |z| < 3$ ; (c3) ROC: |z| > 3.

(d) Let  $x[n] = [\cdots, 0, \boxed{1}, 3, 1, 0, \cdots]$ . Determine the DTFT  $X(e^{j\omega})$ , also determine and give plots of the amplitude spectrum and the phase spectrum.

## Solution

(a) To avoid confusion, write  $r[n] = x[-n] = [\cdots, 0, 2, 3, \boxed{1}, 0, \cdots]$ . The convolution is  $y[n] = \sum_k r[k]x[n-k]$ , where k = -2, -1, 0, hence,

(b) The response is anticausal, stops at n = -2. Shifting to the origin gives  $2^{-n}u[-n]$ , we will need to take into account an 'advance'  $z^2$ .

$$X(z) = z^{2} \sum_{n=-\infty}^{0} (2z)^{-n} = z^{2} \sum_{n=0}^{\infty} (2z)^{n} = \frac{z^{2}}{1-2z}, \qquad \text{ROC: } |z| < \frac{1}{2}.$$

(c) Make proper and do a partial fraction expansion: write as

$$H(z) = A + \frac{B}{1 - \frac{1}{2}z^{-1}} + \frac{C}{1 - 3z^{-1}}$$

where it follows that A = -1, B = 1, C = 1. (c1): Anticausal response. Rewrite

$$H(z) = -1 - \frac{2z}{1 - 2z} - \frac{\frac{1}{3}z}{1 - \frac{1}{3}z}$$

$$h[n] = -\delta[n] - 2(2)^{-n-1}u[-n-1] - \frac{1}{3}\left(\frac{1}{3}\right)^{-n-1}u[-n-1]$$
  
=  $-\delta[n] - 2^{-n}u[-n-1] - 3^nu[-n-1]$ 

(c2): Mixed causality response (2nd term causal, 3rd term anticausal):

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

$$h[n] = -\delta[n] + \left(\frac{1}{2}\right)^n u[n] - \frac{1}{3}\left(\frac{1}{3}\right)^{-n-1} u[-n-1]$$
$$= -\delta[n] + \left(\frac{1}{2}\right)^n u[n] - 3^n u[-n-1]$$

(c3): Causal response:

$$H(z) = -1 + \frac{1}{1 - \frac{1}{2}z^{-1}} + \frac{1}{1 - 3z^{-1}}$$
$$h[n] = -\delta[n] + \left(\frac{1}{2}\right)^n u[n] + 3^n u[n]$$

In each of the above cases, there are several alternative ways to write the answer.

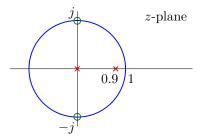
(d)

$$X(e^{j\omega}) = 1 + 3e^{-j\omega} + e^{-j2\omega} = e^{-j\omega}(3 + 2\cos(\omega))$$

Amplitude spectrum:  $|X(e^{j\omega})| = 3 + 2\cos(\omega)$ . Phase spectrum:  $\angle(\omega) = -\omega$ .

### Question 2 (5 points)

Consider the pole-zero plot of a discrete-time causal filter with transfer function H(z):



(a) Determine H(z), up to an amplitude scale factor A.

- (b) Suppose that we know that h[0] = 2. Determine A.
- (c) Based on the pole-zero locations, contruct a sketch of the magnitude spectrum  $|H(e^{j\omega})|$ . Clearly indicate relevant values on the  $\omega$ -axis.
- (d) Specify the ROC. Is this a stable filter?

(a)

$$H(z) = A \frac{(1 - jz^{-1})(1 + jz^{-1})}{(1 - 0z^{-1})(1 - 0.9 z^{-1})} = A \frac{1 + z^{-2}}{1 - 0.9z^{-1}}$$

which is consistent with a pole at zero. Alternatively,

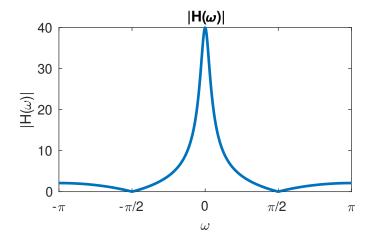
$$H(z) = A \frac{z^2 + 1}{(z - 0.9)z}$$

(b) initial value theorem:

$$h[0] = \lim_{z \to \infty} H(z) = A$$

Hence A = 2.

(c) Use phasors. For the amplitude response, the pole at z = 0 is irrelevant. The pole at z = 1 gives a large peak at  $\omega = 0$ . The zero locations on the unit circle give zero crossings in the amplitude response. At z = -1 ( $\omega = \pm \pi$ ), the response is low. Calculate:  $H(z = 1) = A \cdot 2/0.1 = 40$ ;  $H(z = -1) = A \cdot 2/1.9 \approx 2.1$ . The amplitude response is an even function.

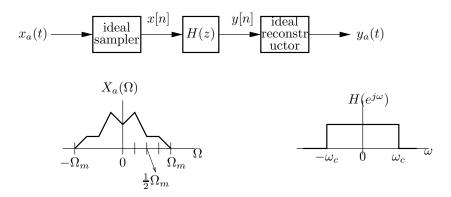


(d) ROC: |z| > 0.9. Stable, because the unit circle is in the ROC (or: all poles are within the unit circle).

### Question 3 (6 points)

A continuous-time signal  $x_a(t)$  has a spectrum  $X_a(\Omega)$  as indicated below. It is sampled at the Nyquist rate (resulting in x[n]), passed through a lowpass filter with frequency response  $H(e^{j\omega})$  (resulting in y[n]), and reconstructed using an ideal DAC (which includes an ideal interpolation filter). The output signal is  $y_a(t)$ .

The cut-off frequency of the lowpass filter is  $\omega_c = \Omega_m T/2$ , where T is the sample period.



- (a) Relate T to  $\Omega_m$ .
- (b) Draw the spectra  $X(\omega)$ ,  $Y(\omega)$ , and  $Y_a(\Omega)$ . Clearly mark the relevant values on the frequency axis.
- (c) Suppose now that we sample at twice the Nyquist rate. Again draw the spectra  $X(\omega)$ ,  $Y(\omega)$ , and  $Y_a(\Omega)$ .

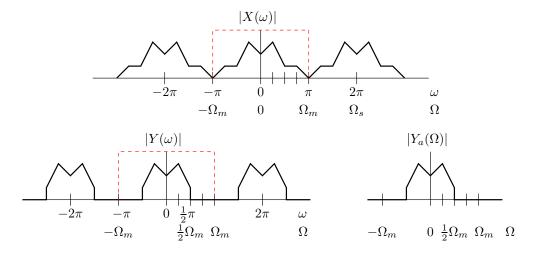
(a) The signal is sampled at Nyquist. Hence,

$$F_s = \frac{1}{T} = \frac{2\Omega_m}{2\pi} \qquad \Rightarrow \qquad T = \frac{\pi}{\Omega_m}$$

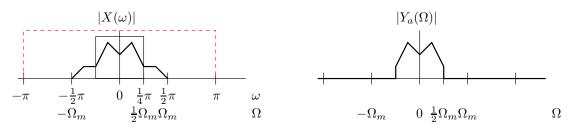
(b)

$$\omega_c = \frac{\Omega_m T}{2} = \frac{1}{2}\pi$$

 $X(\omega)$  is periodic; note on the frequency axis the relation to  $X_a(\Omega)$ .  $Y(\omega)$  is lowpass filtered but also periodic (the red dashed box indicates the fundamental interval).  $Y_a(\Omega)$  is not periodic anymore.



(c) Now,  $\omega_c = \frac{1}{4}\pi$ , but also the mapping  $\Omega \to \omega$  changes:



It follows that the analog signal  $Y_a(\Omega)$  is the same as before.

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# **EE2S11 SIGNALS AND SYSTEMS**

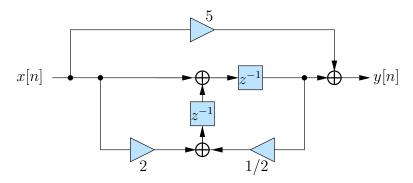
Final exam, 29 January 2021, 13:30–15:50 Block 2 (14:50-15:50)

Open book, strictly timed take-home exam. (Electronic) copies of the book and the course slides allowed. No other tools except a basic pocket calculator permitted. Upload answers during 15:45–16:00

This block consists of three questions (20 points); more than usual, and this will be taken into account during grading. Answer in Dutch or English. Make clear in your answer how you reach the final result; the road to the answer is very important. Write your name and student number on each sheet.

## Question 4 (6 points)

Consider the following realization of a causal system:



- (a) Determine the transfer function H(z).
- (b) What is the difference equation implemented by this realization?
- (c) Is this a stable realization? (motivate)
- (d) Is this a minimal realization? (motivate)
- (e) Draw the "Direct form no. II" realization of the filter and also specify the coefficients.

(a) Let P(z) be the input of the top delay element, then

$$\begin{cases} P(z) &= \frac{1}{2}z^{-2}P(z) + X(z) + 2z^{-1}X(z) \\ Y(z) &= 5X(z) + z^{-1}P(z) \\ \end{cases}$$

$$\begin{cases} P(z) &= \frac{1+2z^{-1}}{1-\frac{1}{2}z^{-2}}X(z) \\ Y(z) &= \left(5+z^{-1}\frac{1+2z^{-1}}{1-\frac{1}{2}z^{-2}}\right)X(z) \\ \end{cases}$$

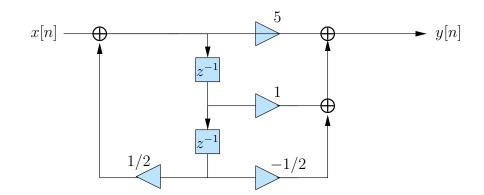
$$H(z) &= \frac{5(1-\frac{1}{2}z^{-2}) + z^{-1}(1+2z^{-1})}{1-\frac{1}{2}z^{-2}} \\ &= \frac{5+z^{-1}-\frac{1}{2}z^{-2}}{1-\frac{1}{2}z^{-2}} \end{cases}$$

(b)

$$y[n] - \frac{1}{2}y[n-2] = 5x[n] + x[n-1] - \frac{1}{2}x[n-2]$$

- (c) Stable, the two poles are  $p_{1,2} = 1/\sqrt{2}$ , within the unit circle.
- (d) Minimal, 2nd order transfer function, and two delays are used.

(e)



## Question 5 (9 points)

A normalized second-order analog low-pass filter (Butterworth filter) is given by

$$H_a(s) = \frac{1}{s^2 + \sqrt{2}\,s + 1}\,.$$

The 3-dB cutoff frequency for this template is  $\Omega_c = 1$  rad/s.

We are asked to design a second-order digital high-pass filter G(z) with

Passband frequency: 
$$\omega_p = 1$$
 rad  
Passband damping: 1 dB  
Stopband frequency:  $\omega_s = 0.5$  rad

We will first design an analog 2nd order high-pass filter  $G_a(s)$  and then apply the bilinear transform.

- (a) From the given specifications, what are the passband and stopband frequencies for the analog high-pass filter?
- (b) Based on  $H_a(s)$ , what is the corresponding power spectrum  $|H_a(j\Omega)|^2$  ?
- (c) What frequency transformation is needed to transform  $|H_a(j\Omega)|^2$  to a template  $|G_a(j\Omega)|^2$  for the analog 2nd order high-pass filter, which involves design parameters  $\epsilon$  and  $\Omega_p$ ?
- (d) What is the corresponding template high-pass filter  $G_a(s)$ ?
- (e) Compute the unknown parameters: What is  $|G_a(j\Omega)|^2$  and  $G_a(s)$  that satisfies the specifications?
- (f) What is the resulting digital high-pass filter G(z) that satisfies the specifications?
- (g) How much damping in the stopband is achieved? (specify in dB)

- (a)  $\Omega_p = \tan(\omega_p/2) = 0.5463, \ \Omega_s = \tan(\omega_s/2) = 0.2553.$
- (b)

$$|H_a(j\Omega)|^2 = H(s)H(-s)|_{s=j\Omega} = \frac{1}{-\Omega^2 + j\sqrt{2}\Omega + 1} \cdot \frac{1}{-\Omega^2 - j\sqrt{2}\Omega + 1} = \frac{1}{(1 - \Omega^2)^2 + 2\Omega^2} = \frac{1}{1 + \Omega^4} + \frac{1}{(1 - \Omega^2)^2 + 2\Omega^2} = \frac{1}{(1 - \Omega^2)^2 + 2\Omega^2}$$

which indeed corresponds to a Butterworth of order 2.

(c) We want to obtain a filter of the form

$$|G_a(j\Omega)|^2 = \frac{1}{1 + \epsilon^2 \left(\frac{\Omega_p}{\Omega}\right)^4}$$

Comparing to (b), the transformation we need is

$$\Omega \to \sqrt{\epsilon} \frac{\Omega_p}{\Omega} \,, \qquad s \to \sqrt{\epsilon} \frac{\Omega_p}{s}$$

(Instead of  $\sqrt{\epsilon}$ , we could use another scale, e.g. introduce a parameter  $\alpha$ , as long as we take that into account into the resulting template for  $|G_a(j\Omega)|^2$ .)

(d) Apply the transformation to  $H_a(s)$ :

$$G_a(s) = \frac{1}{\epsilon \frac{\Omega_p^2}{s^2} + \sqrt{2\epsilon} \frac{\Omega_p}{s} + 1}$$

(e) In the equation for  $|G_a(j\Omega)|^2$ , fill in  $\Omega = \Omega_p$ :

$$|G_a(j\Omega_p)|^2 = \frac{1}{1+\epsilon^2} = 10^{-1/10} = 0.7943$$
$$\epsilon = \sqrt{1/0.7943 - 1} = 0.5089.$$

Hence

$$G_a(s) = \frac{s^2}{0.1519 + 0.5511 \, s + s^2}$$

(f) Substitute the bilinear transform,

$$s \to \frac{1 - z^{-1}}{1 + z^{-1}}$$

resulting in

$$\begin{split} G(z) &= \frac{\frac{(1-z^{-1})^2}{(1+z^{-1})^2}}{0.1519 + 0.5511 \frac{1-z^{-1}}{1+z^{-1}} + \frac{(1-z^{-1})^2}{(1+z^{-1})^2}} \\ &= \frac{(1-z^{-1})^2}{0.1519(1+z^{-1})^2 + 0.5511(1-z^{-1})(1+z^{-1}) + (1-z^{-1})^2} \\ &= \frac{(1-z^{-1})^2}{0.1519 + 2 \cdot 0.1519 z^{-1} + 0.1519 z^{-2} + 0.5511 - 0.5511 z^{-2} + 1 - 2 z^{-1} + z^{-2}} \\ &= \frac{(1-z^{-1})^2}{1.7030 - 1.6962 z^{-1} + 0.6008 z^{-2}} \end{split}$$

(g) Most reliable/straightforward is to fill in  $\Omega_s$  in the formula for  $|G_a(j\Omega)|^2$ :

$$|G_a(j\Omega_s)|^2 = \frac{1}{1 + \epsilon^2 \left(\frac{\Omega_p}{\Omega_s}\right)^4} = \frac{1}{1 + (0.5089)^2 \left(\frac{0.5463}{0.2553}\right)^4} = 0.1555$$

Thus, the damping is  $10 \log(0.1555) = -8.1$  dB.

(You could also take G(z), insert  $z = e^{j\omega_s}$ , and compute the norm of the result. You'll have to deal with complex numbers.)

## Question 6 (5 points)

(a) Determine the Fourier transform of

$$x(t) = \cos(\Omega_0 t) \sin(\Omega_1 t) \,.$$

(b) A periodic signal x(t) has a Fourier series

$$x(t) = \sum_{k=1}^{\infty} \frac{2}{k^2} \cos(3kt/2).$$

Compute the Fourier transform,  $X(\Omega)$ .

(c) Use the duality theorem to prove the following Fourier transform result:

$$x(t) = \frac{1}{t^2 + a^2}, \quad a > 0 \qquad \leftrightarrow \qquad X(\Omega) = \frac{\pi}{a} e^{-a|\Omega|}$$

## Solution

(a) Using the multiplication property,

$$X(\Omega) = \frac{\pi^2}{2\pi} \left[ \delta(\Omega - \Omega_0) + \delta(\Omega + \Omega_0) \right] * (-j) \left[ \delta(\Omega - \Omega_1) - \delta(\Omega + \Omega_1) \right]$$
  
=  $\frac{j\pi}{2} \left[ \delta(\Omega - \Omega_0 + \Omega_1) + \delta(\Omega + \Omega_0 + \Omega_1) - \delta(\Omega - \Omega_0 - \Omega_1) - \delta(\Omega + \Omega_0 - \Omega_1) \right]$ 

(b)

$$X(\Omega) = \pi \sum_{k=1}^{\infty} \frac{2}{k^2} \left[ \delta(\Omega - 3k/2) + \delta(\Omega + 3k/2) \right]$$

(c) Start with the LT pairs

$$\begin{array}{cccc} e^{-at}u(t) & \leftrightarrow & \frac{1}{s+a} \\ e^{at}u(-t) & \leftrightarrow & \frac{1}{-s+a} \\ e^{-a|t|} & \leftrightarrow & \frac{1}{s+a} + \frac{1}{-s+a} = \frac{2a}{a^2-s^2} \end{array}$$

Thus, we have the FT pair  $(s = j\Omega)$ 

$$y(t) = e^{-a|t|} \qquad \leftrightarrow \qquad Y(\Omega) = \frac{2a}{a^2 + \Omega^2}$$

The duality theorem gives then

$$Y(t) = \frac{2a}{a^2 + t^2} \qquad \leftrightarrow \qquad 2\pi y(-\Omega) = 2\pi e^{-a|\Omega|}$$

Finally, by rescaling we obtain then that

$$\frac{1}{a^2 + t^2} \qquad \leftrightarrow \qquad \frac{\pi}{a} e^{-a|\Omega|}$$