# Partial exam EE2S11 Signals and Systems Part 2: trial exam 14 January 2016 (2 hours)

Closed book; two sides of handwritten notes permitted.

This exam consists of five questions (45 points)

## Question 1 (12 points)

Given the signals

$$x[n] = \begin{cases} 1, & 0 \le n \le 5, \\ 0, & \text{elsewhere} \end{cases} \quad h[n] = [\cdots, 0, \ \boxed{1}, \ -2, \ 1, \ 0, \cdots]$$

- a) Determine y[n] = x[n] \* h[n] using the convolution sum (in time-domain).
- b) Determine the z-transforms X(z) and H(z), Also specify the regions of convergence (ROCs).
- c) Determine y[n] = x[n] \* h[n] using the (inverse) z-transform.
- d) Given

$$X(z) = \frac{1}{1 - 1\frac{1}{2}z^{-1} + \frac{1}{2}z^{-2}}, \qquad z \in \text{ROC}.$$

Determine x[n] using the inverse z-transform if (d1) ROC: |z| > 1, (d2) ROC:  $|z| < \frac{1}{2}$ , (d3) ROC:  $\frac{1}{2} < |z| < 1$ .

- e) Given  $x[n] = (-1)^n u[n]$ . Determine the DTFT  $X(\omega)$ .
- f) Given  $X(\omega) = \cos(\omega)$ . Determine x[n].

### Question 2 (8 points)

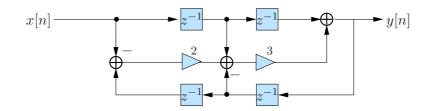
The response of a causal LTI system is given by

$$H(z) = \frac{z+1}{(z+1)^2+1}, \quad z \in \text{ROC}$$

- a) Determine all poles and zeros of the system, and draw them in the complex z-plane.
- b) Specify the ROC.
- c) Is the system BIBO stable? (Why?)
- d) Determine the frequency response of the system.
- e) Determine the impulse response h[n].

### Question 3 (8 points)

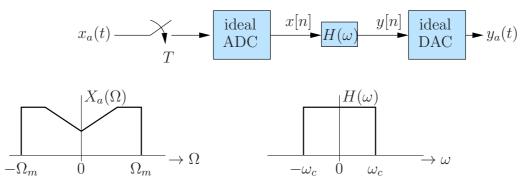
Given the following system:



- a) Determine the transfer functions H(z) of the system.
- b) Is this a stable system? (Why?)
- c) Is this a minimal realization? (Why?)
- d) Draw the "Direct Form no. II" realization, and also specify the coefficients.

#### Question 4 (7 points)

A continuous-time signal  $x_a(t)$  has a spectrum as indicated below.  $x_a(t)$  is sampled with a frequency 1/T equal to the Nyquist rate, filtered by an ideal low-pass filter  $H(\omega)$ , and converted back to an analog signal  $y_a(t)$ . The cut-off frequency of  $H(\omega)$  is  $\omega_c = \Omega_m T/3$ .



- a) Give an expression for T.
- b) Draw the spectrum corresponding to x[n]. Also mark the frequencies.
- c) Draw the spectrum corresponding to y[n]. Also mark the frequencies.
- d) Draw the spectrum corresponding to  $y_a(t)$ . Also mark the frequencies.

#### Question 5 (10 points)

We would like to design a digital low-pass filter with the following specifications:

- Ripple in the pass-band :  $\leq 1 \text{ dB}$
- Pass-band: 4 kHz
- Stop-band damping:  $\geq$  40 dB
- Stop-band: starting at 6.0 kHz
- Sample rate: 24 kHz

The digital filter is designed by applying the bilinear transform to an analog transfer function.

- a) What are the pass-band and stop-band frequencies in the digital time-domain?
- b) What are the filter specifications in the analog time-domain?
- c) Compute the required filter order for a Butterworth filter

- d) Compute the required filter order for a Chebyshev filter (Remark:  $\cosh^{-1}(x) = \ln(x + \sqrt{x^2 1}).)$
- e) Draw the frequency response of the resulting two digital filters after applying the bilinear transform. Also indicate the relation to the filter specifications.