

COM-303 - Signal Processing for Communications Final Exam

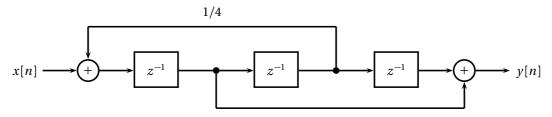
July 2, 2015, 08:15 to 11:15

Verify that this exam has YOUR last name on top DO NOT OPEN THE EXAM UNTIL INSTRUCTED TO DO SO

- Write your name on the top left corner of ALL the sheets you turn in.
- There are 4 problems for a total of 100 points; the number of points is indicated for each problem.
- Please write your derivations clearly!
- You can have two A4 sheets of *handwritten* notes (front and back). Please **no photocopies, no books and no electronic devices**. Turn off your phone and store it in your bag.
- When you are done, simply leave your solution on your desk **with this page on top** and exit the classroom.

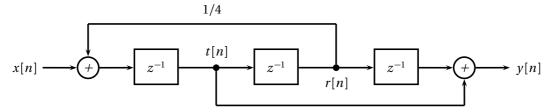
Exercise 1. (25 points)

Consider the causal system implemented by the following block diagram:



- (a) Compute the system's transfer function H(z)
- (b) Plot the system's poles and zeros on the complex plane
- (c) Sketch the magnitude of the system's frequency response $|H(e^{j\omega})|$
- (d) Draw another block diagram that implements the same transfer function H(z) as a cascade of a second-order direct form II structure and a simple delay.

First set two auxiliary variables t[n] and r[n] like so:



2

(a) From simple inspection:

$$y[n] = t[n] + r[n-1]$$

 $r[n] = t[n-1]$
 $t[n] = x[n-1] + (1/4)r[n-1]$

which, in the z-domain, becomes

$$Y(z) = T(z) + z^{-1}R(z)$$

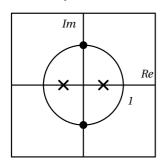
$$R(z) = z^{-1}T(z)$$

$$T(z) = z^{-1}X(z) + (1/4)z^{-1}R(z)$$

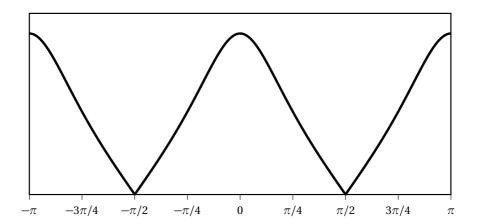
Solving for X(z) and Y(z) yields

$$H(z) = \frac{X(z)}{Y(z)} = z^{-1} \frac{1 + z^{-2}}{1 - (1/4)z^{-2}}$$

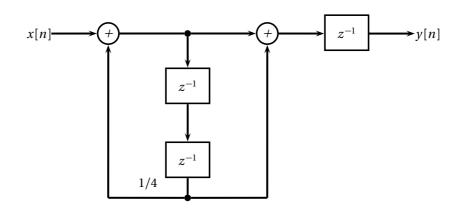
(b) the roots of the numerator are $\pm j$ and those of the denominator are $\pm 1/2$ therefore



(c) The filter is an approximate stopband



(*d*)



Exercise 2. (25 points)

Plot the DTFT of the signal x[n] = sinc(5n/2).

Hint: you can either work mostly in the time domain using simple trigonometry (but careful with the value of x[0]) or you can work mostly in the frequency domain by considering x[n] as a continuous-time sinc sampled with period $T_s = 5/2$; in this case the grid below can be of help.

$$-2\pi - 9\pi/5 - 8\pi/5 - 7\pi/5 - 6\pi/5 - \pi - 4\pi/5 - 3\pi/5 - 2\pi/5 - \pi/5 = 0$$

$$\pi/5 = 2\pi/5 - 3\pi/5 - 4\pi/5 - \pi/5 = 0$$

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The standard derivation of the rect-sinc DTFT pair starts from a rect in frequency. Since the cutoff of the rect must be less than π , the inverse DTFT returns a signal of the form $\operatorname{sinc}(\alpha n)$ with $\alpha < 1$ and therefore we cannot use the rect-sinc pair formula if, like in this case, $\alpha > 1$.

Working in the time domain: by exploting the 2π -periodicity of the sine, we have for $n \neq 0$:

$$\operatorname{sinc}(5n/2) = \frac{\sin 5\pi n/2}{5\pi n/2} = \frac{\sin(2\pi n + \pi n/2)}{5\pi n/2} = (1/5)\frac{\sin \pi n/2}{\pi n/2} = (1/5)\operatorname{sinc}(n/2)$$

For n = 0, x[n] = 1 while the above expression is equal to 1/5. Therefore we can write

$$x[n] = (1/5)\operatorname{sinc}(n/2) + (4/5)\delta[n]$$

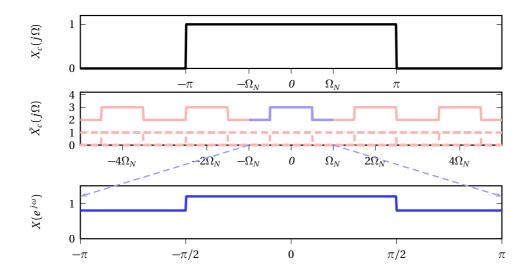
We can now use the standard sinc-rect transform pair formula to obtain

$$X(e^{j\omega}) = 4/5 + (2/5)\text{rect}(\omega/\pi)$$

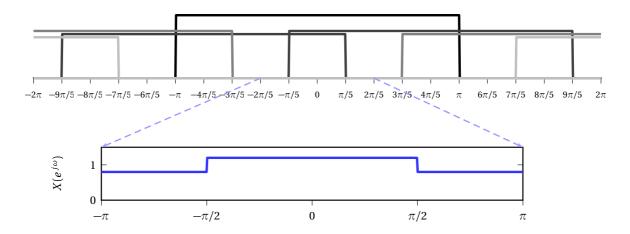
Working in the frequency domain: Consider the continuous-time sinc function $x(t) = \operatorname{sinc}(t)$; its Fourier transform is $\Phi(j\Omega) = \operatorname{rect}(\Omega/2\pi)$, bandlimited to $\Omega_0 = \pi$. Clearly x[n] = x(nT) for $T_s = 5/2$; however, 5/2 is larger than the maximum alias-free sampling period for x(t), which is $T_{\max} = \pi/\Omega_0 = 1$ so we will have aliasing. The resulting DTFT will be

$$X(e^{j\omega}) = \frac{1}{T_s} \sum_{k=-\infty}^{\infty} \Phi\left(\frac{j\omega}{\pi} \Omega_N + 2jk\Omega_N\right)$$

with $\Omega_N = \pi/T_s = 2\pi/5$. We can determine the shape of the DTFT graphically:



The overlaps are illustrated in detail by this figure, using an artificial different height for each spectral copy:

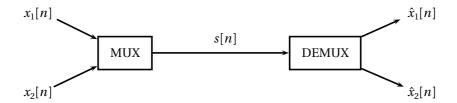


so that

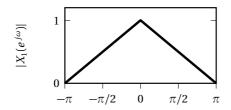
$$X(e^{j\omega}) = \begin{cases} 6/5 & for |\omega| < \pi/2 \\ 4/5 & otherwise \end{cases}$$
 extended by 2π -periodicity

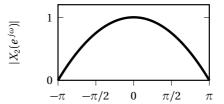
Exercise 3. (26 points)

In communication systems, it often happens that several data streams must share a single communication channel; in these cases the data streams are *multiplexed* and transmitted together and are de-multiplexed at the receiving end:



The two simplest multiplexing schemes are time-division multiplexing (TDM) and frequency-division multiplexing (FDM). Consider the simple case of two discrete-time data sequences $x_1[n]$ and $x_2[n]$, whose magnitude spectra are sketched here:

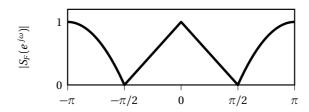




Using 2-TDM (i.e. time-division multiplexing for two streams), the transmitted sequence $s_T[n]$ is obtained by interleaving the original sequences:

$$s_T[n] = \begin{cases} x_1[n/2] & \text{for } n \text{ even} \\ x_2[(n-1)/2] & \text{for } n \text{ odd} \end{cases}$$

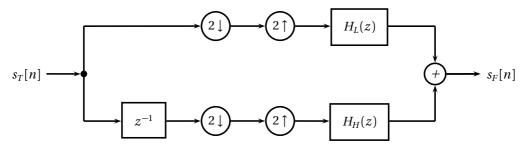
In 2-FDM, on the other hand, the two sequences are combined into a sequence $s_F[n]$ so that $S_F(e^{j\omega})$ looks like this:



- (a) Assume you receive a 2-TDM signal and you need to retransmit it to a 2-FDM receiver. Draw the block diagram of a system that takes $s_T[n]$ as the input and converts it into $s_F[n]$ as the output. There should be no data loss and the data rate (i.e. number of samples per second) should remain the same between the input and the output.
- (b) Design an 2-FDM de-multiplexer (i.e. a system that takes $s_F[n]$ as the input and produces the original components $x_1[n]$ and $x_2[n]$).

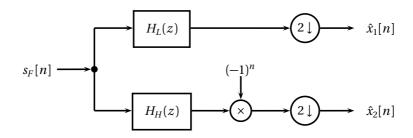
Please be extremely precise in your block diagrams.

(a)



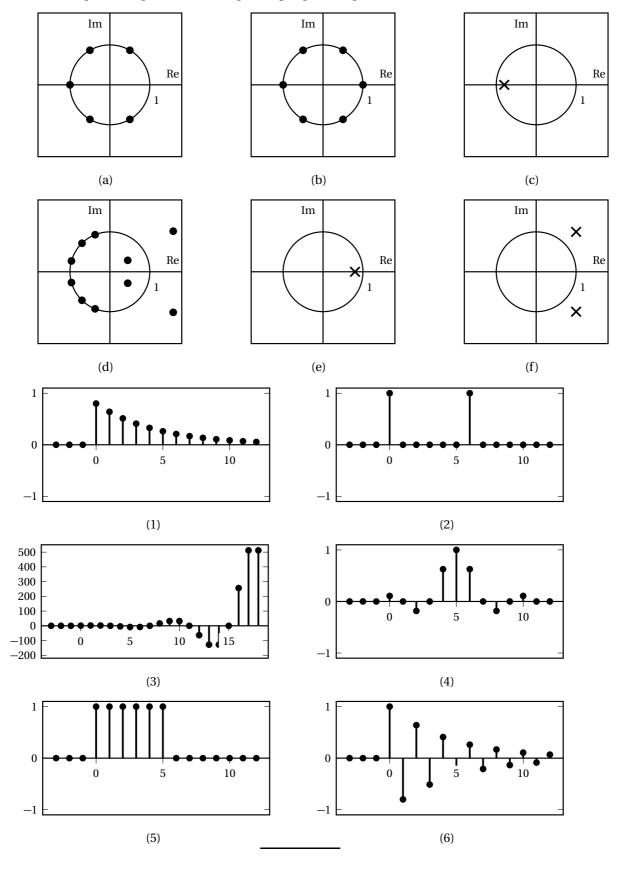
where $H_L(z)$ and $H_H(z)$ are ideal low- and high-pass filters with cutoff frequency $\pi/2$.

(b)



Exercise 4. (24 points)

Associate each pole-zero plot to the corresponding impulse response.



- (a) (5)
- (b) (2)
- (c) (6) (d) (4)
- (e) (1)
- (f) (3)