

**Final Exam**

8:00-11:00 AM, Wednesday, May 12, 2021

**Please do not start the exam until the starting time.**

- You may not use any books, electronic devices, or notes other than **four** handwritten two-sided sheet of 8.5" x 11" paper.
- You should solve the problems on blank sheets of paper, take pictures of your solutions, and upload them to Gradescope before the end of the exam time.

**GOOD LUCK!**

1. (10 Pts.) Answer **True** or **False** to each of the following statements:

- (a) The phase response of a real-valued filter has odd symmetry. **T/F**
- (b) Strictly linear phase filters are always GLP filters. **T/F**
- (c) Applying transposition to a filter structure with transfer function  $H(z)$  may change  $H(z)$ . **T/F**
- (d) If a lowpass filter is designed using the window FIR filter design method, same errors (ripples) will be achieved in both passband and stopband. **T/F**
- (e) In spectral analysis, to improve the ability to distinguish between two closely spaced frequencies, one should use more samples of the signal because a longer window has a narrower main lobe in the frequency domain. **T/F**
- (f) The upsampling operation

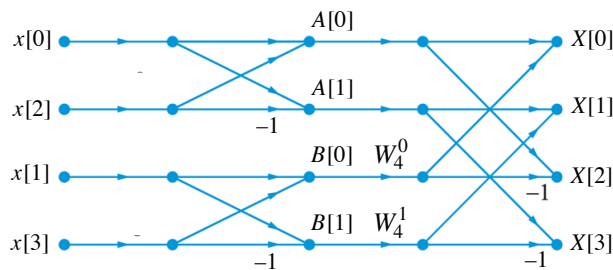
$$y[n] = \begin{cases} x[n/3] & \text{if } n = 3k \text{ for some integer } k \\ 0 & \text{otherwise} \end{cases}$$

for  $-\infty < n < \infty$ , is linear.

- (g) Downsampling can cause aliasing. **T/F**
- (h) Type IV GLP FIR filters can be used to implement lowpass filters. **T/F**
- (i) Consider a radix-2 decimation in time FFT with size  $N = 32$ . Then the FFT will have five stages of computation. **T/F**
- (j) Direct Form II implementations require fewer multiplications per output sample than Direct Form I implementations. **T/F**

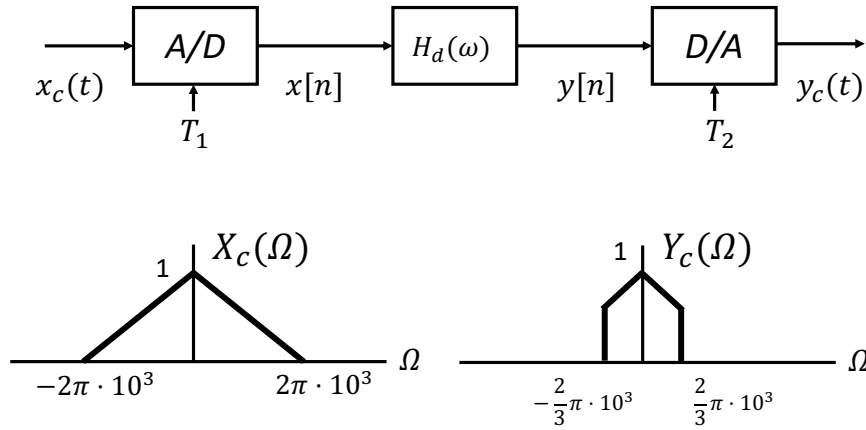
2. (6 Pts.) A real continuous-time signal  $x_c(t)$  is bandlimited to frequencies below 8kHz, i.e.,  $X_c(\Omega) = 0$  for  $|\Omega| \geq 2\pi(8000)$ . Suppose that  $x_c(t)$  is sampled with sampling period  $T_s = 8 \cdot 10^{-5}$ s to produce  $x[n] = x_c(nT_s)$  and 4000 samples are extracted corresponding to  $n = 0, 1, \dots, 3999$ . Let  $\{X[k]\}_{k=0}^{3999}$  be the 4000-point DFT of  $\{x[n]\}_{n=0}^{3999}$ . To what continuous-time frequencies (in Hz) do the indices  $k = 500$  and  $k = 2800$  correspond?

3. (4 Pts.) You would like to compute the DFT of  $\{x[n]\}_{n=0}^3 = \{3, 3, 1, 7\}$  using a radix-2 decimation-in-time FFT, with the following flow graph:



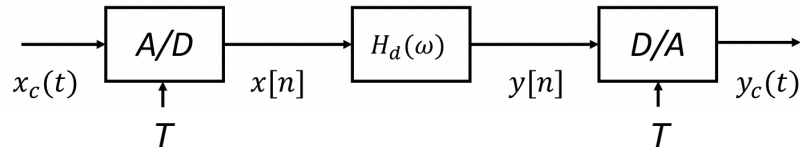
Find the values of  $A[0], A[1], B[0], B[1], X[0], X[1], X[2], X[3]$ .

4. (8 Pts.) Consider the following system with an ideal A/D and an ideal D/A. The input signal  $x_c(t)$  and output signal  $y_c(t)$  have the continuous-time Fourier transforms shown in the figure below.



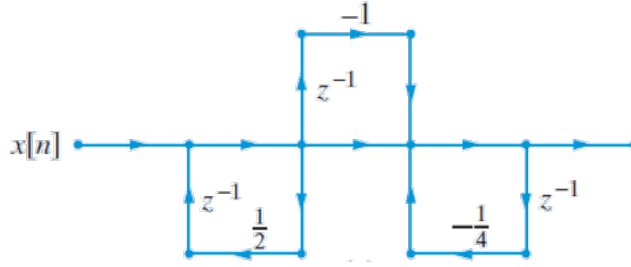
Sketch the desired frequency response of the digital filter  $H_d(\omega)$  for this system for each of the following cases.

- (a)  $T_1 = T_2 = \frac{1}{2 \cdot 10^3}$ .  
 (b)  $T_1 = \frac{1}{2 \cdot 10^3}$  and  $T_2 = \frac{1}{10^3}$
5. (6 Pts.) A speech signal  $x_c(t)$  is assumed to be bandlimited to 30 kHz. You would like to use the following setup in order filter this signal:



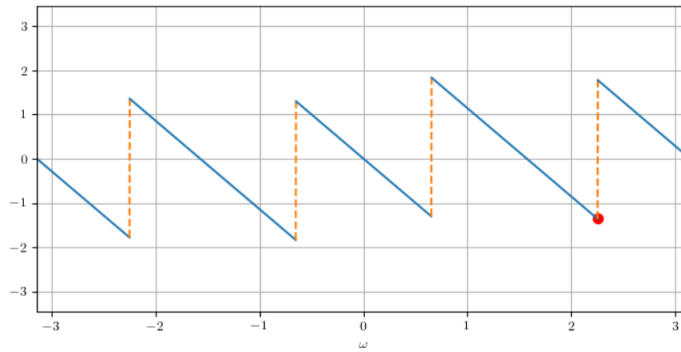
You would like to pass all frequencies between 7.5 kHz and 15 kHz, while suppressing all other frequencies. Determine the maximum sampling rate  $T$  that can be used to perform this task and sketch the frequency response  $H_d(\omega)$  of the ideal filter needed.

6. (8 Pts.) A discrete-time system is described by the following signal flow graph:



- Determine the transfer function of the filter.
- Apply transposition in order to obtain an equivalent filter structure.

7. (6 Pts.) Consider the following phase response  $\angle H_d(\omega)$  of an FIR filter with  $\omega \in [-\pi, \pi]$ :



- Has this filter strict linear phase? Justify your answer.
- Has this filter (a) even symmetry, (b) odd symmetry, (c) neither even nor odd symmetry. Justify your answer.
- Determine  $H_d(\omega_0)$  at the red point in the plot.

8. (5 Pts.) Let  $x[n]$  be a sequence that is zero outside the interval  $[0, N-1]$  with DTFT  $X_d(\omega)$ . Define the signals

$$y[n] = \begin{cases} x[n], & n = 0, 1, \dots, N-1 \\ 0, & n = N, \dots, 3N-1 \end{cases} \quad \text{and} \quad g[n] = \begin{cases} x[\frac{n}{3}], & n = 0, 3, 6, \dots \\ 0, & \text{otherwise} \end{cases}, n = 0, 1, \dots, 3N-1.$$

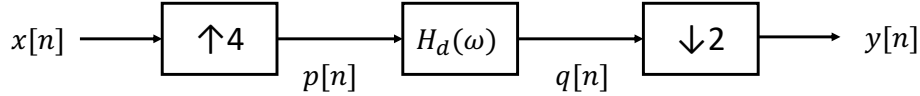
Let  $Y[k]$  and  $G[k]$  be the  $3N$ -point DFTs of  $y[n]$  and  $g[n]$ . Select the correct expression for the DFT of each sequence in terms of samples of  $X_d(\omega)$ .

- $Y[k] = X_d\left(\frac{2\pi}{N}k\right)$  and  $G[k] = X_d\left(\frac{2\pi}{3N}k\right)$  for  $k = 0, 1, \dots, 3N-1$
- $Y[k] = X_d\left(\frac{2\pi}{3N}k\right)$  and  $G[k] = X_d\left(\frac{6\pi}{N}k\right)$  for  $k = 0, 1, \dots, 3N-1$
- $Y[k] = X_d\left(\frac{2\pi}{3N}k\right)$  and  $G[k] = X_d\left(\frac{2\pi}{N}k\right)$  for  $k = 0, 1, \dots, 3N-1$
- $Y[k] = X_d\left(\frac{2\pi}{3N}k\right)$  and  $G[k] = \begin{cases} X_d\left(\frac{2\pi}{N}k\right), & k = 0, 3, 6, \dots \\ 0, & \text{otherwise} \end{cases}, k = 0, 1, \dots, 3N-1$
- None of the above

9. (8 Pts.) Let  $\{x[n]\}_{n=0}^3 = \{1, 3, -2, 4\}$  and  $\{h[n]\}_{n=0}^2 = \{-1, 1, 3\}$ .

- Compute the circular convolution  $x[n] \otimes_4 h_{zp}[n]$  where  $\{h_{zp}[n]\}_{n=0}^3 = \{h[0], h[1], h[2], 0\}$ .
- Compute the linear convolution  $x[n] * h[n]$ .
- What is the smallest value of  $N$  so that  $N$ -point circular convolution is equal to the linear convolution?
- If you would like to compute the linear convolution  $x[n] * h[n]$  using the DFT method employing the radix-2 decimation-in-time FFT algorithm, how many zeros should be padded to  $x[n]$  and  $h[n]$ , respectively?

10. (9 Pts.) Consider the system in the figure below.



where the frequency response of  $H_d(\omega)$  is

$$H_d(\omega) = \begin{cases} 1, & \frac{1}{4}\pi \leq |\omega| \leq \frac{1}{2}\pi \\ 0, & \text{otherwise.} \end{cases}$$

Suppose that an input  $x[n]$  has the following DTFT

$$X_d(\omega) = \begin{cases} 1, & 0 < \omega \leq \pi \\ \frac{\omega}{\pi} + 1, & -\pi < \omega \leq 0. \end{cases}$$

- Sketch  $P_d(\omega)$ .
- Sketch  $Q_d(\omega)$ .
- Sketch  $Y_d(\omega)$ .

11. (8 Pts.)

- Determine the unit impulse response  $h_1[n]$  of an ideal discrete-time lowpass filter with cutoff frequency  $\omega_c = \pi/5$ .
- Determine the unit impulse response  $h_2[n]$  of an ideal discrete-time bandpass filter satisfying the following specification:

$$H_d(\omega) = \begin{cases} 1, & \frac{3}{10}\pi \leq |\omega| \leq \frac{7}{10}\pi \\ 0, & \text{otherwise.} \end{cases}$$

**Hint:** Use the result from (a) and the DTFT pair,

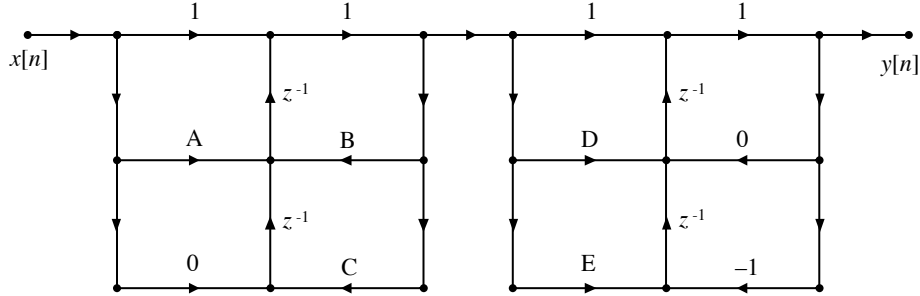
$$\cos(\omega_0 n) \cdot x[n] \longleftrightarrow \frac{1}{2}X_d(\omega - \omega_0) + \frac{1}{2}X_d(\omega + \omega_0).$$

- Determine a length-7 causal GLP FIR filter  $g[n]$  that approximates the above desired bandpass filter  $h_2[n]$  using a rectangular window design.

12. (10 Pts.) Consider a causal LTI system with the following transfer function:

$$H(z) = \frac{(1 - z^{-3})}{(1 - jz^{-1})(1 - \frac{1}{4}z^{-1})(1 - \frac{1}{3}z^{-1})(1 + jz^{-1})}$$

You would like to implement it in cascade form using second-order transpose form II sections, according to the flow graph shown below. Determine the value of A, B, C, D and E.



13. (6 Pts.) The impulse response of a filter is given by

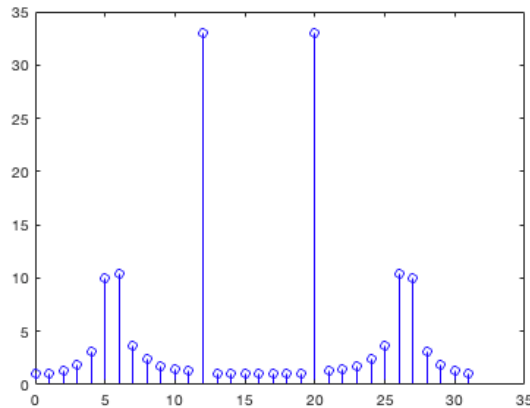
$$h[n] = \delta[n] + 3\delta[n - 1] + \delta[n - 2].$$

- Sketch the magnitude and phase of the filter's frequency response.
- Does the filter have strictly linear phase? Justify your answer.
- What is the output  $y[n]$  when the input is  $x[n] = 5 + e^{j\pi n/4}$  for all  $n$ ?

14. (6 Pts.) Suppose we have a discrete-time signal  $x[n]$  defined as

$$x[n] = A_1 \cos\left(\frac{k_1\pi}{32}n\right) + A_2 \cos\left(\frac{3\pi}{4}n\right),$$

where  $A_1$ ,  $A_2$  and  $k_1$  are known to be positive integers. You compute the 32-point DFT of  $\{x[n]\}_{n=0}^{31}$  and observe that the magnitude of  $\{X[k]\}_{k=0}^{31}$  is as shown below:



Find the values of  $A_1$ ,  $A_2$  and  $k_1$ .