

Name:

**ECE301 Signals and Systems Thursday, April 18, 2019**

**Exam 3**

## Cover Sheet

Write your name on this page and every page to be safe.

Test Duration: 80 minutes.

Coverage: Comprehensive

Open Book but Closed Notes. Three two-sided handwritten sheets.

Calculators NOT allowed.

This test contains **two** problems, each with multiple parts.

You have to draw your own plots.

You must show all work for each problem to receive full credit.

**Problem 1** Consider an analog signal  $x_a(t)$  below with maximum frequency  $\omega_M = 40$  rads/sec: the Fourier Transform,  $X_a(\omega)$ , of  $x_a(t)$  is exactly zero for  $|\omega| > 40$  rads/sec.

$$x_a(t) = \left\{ \frac{\pi \sin(5t)}{5} \frac{\sin(15t)}{\pi t} \right\} 2j \sin(20t)$$

This is the same signal that is sampled for each and every part of entire Problem 1.

**VIP:** For EACH part, you are required to plot the DTFT of  $h(t)$  which is the lowpass filter impulse response that is used for interpolation and determine (and state) whether it is flat over the band  $-\omega_M < \omega < \omega_M$  and whether it filters out all the spectral replicas outside of the band  $-\omega_M < \omega < \omega_M$ , where  $\omega_M = 40$  rads/sec for all parts.

**Problem 1 (a).** The signal  $x_a(t)$  is sampled at a rate  $\omega_s = 100$  rads/sec., where  $\omega_s = 2\pi/T_s$  such the time between samples is  $T_s = \frac{2\pi}{100}$  sec. This yields the discrete-time sequence

$$x[n] = x_a(nT_s) = \left\{ \frac{\pi \sin\left(\frac{\pi}{10}n\right)}{5} \frac{\sin\left(\frac{3\pi}{10}n\right)}{\pi nT_s} \right\} 2j \sin\left(\frac{2\pi}{5}n\right) \quad \text{where: } T_s = \frac{2\pi}{100}$$

A reconstructed signal is formed from the samples above according to the formula below. Determine a simple, closed-form expression for the reconstructed signal  $x_r(t)$ . Show work.

$$x_r(t) = \sum_{n=-\infty}^{\infty} x[n]h(t - nT_s) \quad \text{where: } T_s = \frac{2\pi}{100} \quad \text{and} \quad h(t) = T_s \frac{\pi \sin(10t)}{10} \frac{\sin(50t)}{\pi t}$$

**Problem 1 (b).** Consider the SAME analog signal  $x_a(t)$  with maximum frequency (bandwidth)  $\omega_M = 40$  rads/sec. This signal is sampled at the same rate  $\omega_s = 100$  rads/sec., but is reconstructed with a different lowpass interpolating filter according to the formula below. Does this achieve perfect reconstruction, that is, does  $x_r(t) = x_a(t)$ ? **For this part, you do not need to determine  $x_r(t)$ , just need to explain whether  $x_r(t) = x_a(t)$  or not.**

$$x_r(t) = \sum_{n=-\infty}^{\infty} x[n]h(t - nT_s) \quad \text{where: } T_s = \frac{2\pi}{100} \quad \text{and} \quad h(t) = T_s \frac{1}{2} \left\{ \frac{\sin(40t)}{\pi t} + \frac{\sin(60t)}{\pi t} \right\}$$

**Problem 1 (c).** Consider the SAME analog signal  $x_a(t)$  with maximum frequency (bandwidth)  $\omega_M = 40$  rads/sec. This signal is sampled at the same rate  $\omega_s = 100$  rads/sec., but is reconstructed with a different lowpass interpolating filter according to the formula below. Does this achieve perfect reconstruction, that is, does  $x_r(t) = x_a(t)$ ? **For this part, you do not need to determine  $x_r(t)$ , just need to explain whether  $x_r(t) = x_a(t)$  or not.**

$$x_r(t) = \sum_{n=-\infty}^{\infty} x[n]h(t - nT_s) \quad \text{where: } T_s = \frac{2\pi}{100} \quad \text{and} \quad h(t) = T_s \frac{1}{2} \left\{ \frac{\sin(45t)}{\pi t} + \frac{\sin(55t)}{\pi t} \right\}$$

Show all your work for Prob. 1, parts (a)-(b)-(c) .

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**Problem 1 (d).** Consider an analog signal  $x_a(t)$  with maximum frequency (bandwidth)  $\omega_M = 40$  rads/sec. That is, the Fourier Transform of the analog signal  $x_a(t)$  is exactly zero for  $|\omega| > 40$  rads/sec. This signal is sampled at a rate  $\omega_s = 80$  rads/sec., where  $\omega_s = 2\pi/T_s$  such the time between samples is  $T_s = \frac{2\pi}{80}$  sec. This yields the discrete-time sequence

$$x[n] = x_a(nT_s) = \left\{ \frac{\pi \sin\left(\frac{\pi}{8}n\right) \sin\left(\frac{3\pi}{8}n\right)}{5 \pi n T_s} \right\} 2j \sin\left(\frac{\pi}{2}n\right) \quad \text{where: } T_s = \frac{2\pi}{80}$$

A reconstructed signal is formed from the samples above according to the formula below. Determine a simple, closed-form expression for the reconstructed signal  $x_r(t)$ . Show work.

$$x_r(t) = \sum_{n=-\infty}^{\infty} x[n]h(t - nT_s) \quad \text{where: } T_s = \frac{2\pi}{80} \quad \text{and} \quad h(t) = T_s \frac{\sin(40t)}{\pi t}$$

**Problem 1 (e).** Consider the SAME analog signal  $x_a(t)$  with maximum frequency (bandwidth)  $\omega_M = 40$  rads/sec. This signal is sampled at the same rate  $\omega_s = 80$  rads/sec., but reconstructed with a different lowpass interpolating filter according to the formula below. Does this achieve perfect reconstruction, that is, does  $x_r(t) = x_a(t)$ ? **For this part, you do not need to determine  $x_r(t)$ , just need to explain whether  $x_r(t) = x_a(t)$  or not.**

$$x_r(t) = \sum_{n=-\infty}^{\infty} x[n]h(t - nT_s) \quad \text{where: } T_s = \frac{2\pi}{80} \quad \text{and} \quad h(t) = T_s \left\{ \frac{\pi \sin(5t) \sin(40t)}{5 \pi t \pi t} \right\}$$

**Problem 1 (f).** Consider the SAME analog signal  $x_a(t)$  with maximum frequency (bandwidth)  $\omega_M = 40$  rads/sec. This signal is sampled at the same rate  $\omega_s = 80$  rads/sec., but reconstructed with a different lowpass interpolating filter according to the formula below. Does this achieve perfect reconstruction, that is, does  $x_r(t) = x_a(t)$ ? **For this part, you do not need to determine  $x_r(t)$ , just need to explain whether  $x_r(t) = x_a(t)$  or not.**

$$x_r(t) = \sum_{n=-\infty}^{\infty} x[n]h(t - nT_s) \quad \text{where: } T_s = \frac{2\pi}{80} \quad \text{and} \quad h(t) = T_s \frac{1}{2} \left\{ \frac{\sin(40t)}{\pi t} + \frac{\sin(10t)}{\pi t} \right\}$$

**Show all your work for Prob. 1, parts (d)-(e)-(f).**

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**Problem 1 (g).** Consider an analog signal  $x_a(t)$  with maximum frequency (bandwidth)  $\omega_M = 40$  rads/sec. This signal is sampled at a rate  $\omega_s = 50$  rads/sec., where  $\omega_s = 2\pi/T_s$  such the time between samples is  $T_s = \frac{2\pi}{50}$  sec, yielding the following discrete-time sequence:

$$x[n] = x_a(nT_s) = \left\{ \frac{\pi \sin\left(\frac{\pi}{5}n\right) \sin\left(\frac{3\pi}{5}n\right)}{5 \pi n T_s} \right\} 2j \sin\left(\frac{4\pi}{5}n\right) \quad \text{where: } T_s = \frac{2\pi}{50}$$

A reconstructed signal is formed from the samples above according to the formula below. Determine a closed-form expression for the reconstructed signal  $x_r(t)$ . Show all work.

$$x_r(t) = \sum_{n=-\infty}^{\infty} x[n]h(t - nT_s) \quad \text{where: } T_s = \frac{2\pi}{50} \quad \text{and} \quad h(t) = T_s \frac{\sin(25t)}{\pi t}$$

**Problem 1 (h).** Consider the SAME analog signal  $x_a(t)$  with maximum frequency (bandwidth)  $\omega_M = 40$  rads/sec. This signal is sampled at the same rate  $\omega_s = 50$  rads/sec., where  $\omega_s = 2\pi/T_s$  and the time between samples is  $T_s = \frac{2\pi}{50}$  sec, but at a different starting point. This yields the Discrete-Time  $x[n]$  signal below:

$$x_\epsilon[n] = x_a(nT_s + 0.5T_s) = \left\{ \frac{\pi}{5} \right\} \left\{ \frac{\sin\left(\frac{\pi}{5}(n + 0.5)\right) \sin\left(\frac{3\pi}{5}(n + 0.5)\right)}{\pi(n + 0.5)T_s} \right\} 2j \sin\left(\frac{4\pi}{5}(n + 0.5)\right)$$

A reconstructed signal is formed from the samples above according to the formula below. Determine a simple, closed-form expression for the reconstructed signal  $x_r(t)$ . *Hint:* before you do a lot of work, look at the interpolating lowpass filter being used below.

$$x_r(t) = \sum_{n=-\infty}^{\infty} x_\epsilon[n]h(t - (n + 0.5)T_s) \quad \text{where: } T_s = \frac{2\pi}{50} \quad \text{and} \quad h(t) = T_s \frac{\sin(25t)}{\pi t}$$

**Show your work for Prob. 1, parts (g)-(h) .**

Show all your work for Prob. 1, parts (g)-(h) on this page.

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**Problem 2.** Consider the input signal  $x_0(t)$  below.

$$x_0(t) = e^{-j32t} + e^{-j28t} + e^{-j24t} + e^{-j16t} + e^{-j12t} + e^{-j8t} + e^{-j4t} + 1 + e^{j4t} + e^{j8t} + e^{j12t} + e^{j16t} + e^{j24t} + e^{j28t} + e^{j32t}$$

This signal is first input to an analog filter with impulse response

$$h_{LP}(t) = 3 \frac{\sin(5t)}{\pi t} 2j \sin(5t) + 2 \frac{\sin(5t)}{\pi t} 2j \sin(15t) + \frac{\sin(5t)}{\pi t} 2j \sin(25t)$$

to form  $x(t) = x_0(t) * h_{LP}(t)$ , and then  $x(t)$  is sampled at a rate of  $\omega_s = 64$  to form  $x[n]$ , so that the time between samples is  $T_s = \frac{2\pi}{64}$ . The DT signal  $x[n]$  thus obtained is then input to a DT LTI system with impulse response

$$h[n] = 16 \left\{ \frac{\sin\left(\frac{\pi}{4}n\right)}{\pi n} \right\}^2 2j \sin\left(\frac{\pi}{2}n\right) \quad (1)$$

Show all work. Write your expression for the output  $y[n] = x[n] * h[n]$  in the space below. Plot both the Fourier Transform of  $h_{LP}(t)$  and the DTFT of  $h[n]$  to help solve the problem, but since the input is a sum of sinewaves, it is not necessary to plot the Fourier Transform of  $x_0(t)$  or the DTFT of the sampled signal  $x[n]$ .

Problem 2. You can continue your work for 2 here.

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