# Name: <br> Exam 3 <br> ECE301 Signals and Systems Thursday, April 18, 2019 

## 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 $/ \mathrm{sec}$ : the Fourier Transform, $X_{a}(\omega)$, of $x_{a}(t)$ is exactly zero for $|\omega|>40 \mathrm{rads} / \mathrm{sec}$.

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

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 $\omega_{M}=40 \mathrm{rads} / \mathrm{sec}$ for all parts.

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

$$
x[n]=x_{a}\left(n T_{s}\right)=\left\{\frac{\pi}{5} \frac{\sin \left(\frac{\pi}{10} n\right)}{\pi n T_{s}} \frac{\sin \left(\frac{3 \pi}{10} n\right)}{\pi n T_{s}}\right\} 2 j \sin \left(\frac{2 \pi}{5} n\right) \quad \text { where: } \quad T_{s}=\frac{2 \pi}{100}
$$

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.

$$
x_{r}(t)=\sum_{n=-\infty}^{\infty} x[n] h\left(t-n T_{s}\right) \quad \text { where: } \quad T_{s}=\frac{2 \pi}{100} \quad \text { and } \quad h(t)=T_{s} \frac{\pi}{10} \frac{\sin (10 t)}{\pi t} \frac{\sin (50 t)}{\pi t}
$$

Problem 1 (b). Consider the SAME analog signal $x_{a}(t)$ with maximum frequency (bandwidth) $\omega_{M}=40 \mathrm{rads} / \mathrm{sec}$. This signal is sampled at the same rate $\omega_{s}=100 \mathrm{rads} / \mathrm{sec}$., 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. $x_{r}(t)=\sum_{n=-\infty}^{\infty} x[n] h\left(t-n T_{s}\right) \quad$ where: $\quad T_{s}=\frac{2 \pi}{100} \quad$ and $\quad h(t)=T_{s} \frac{1}{2}\left\{\frac{\sin (40 t)}{\pi t}+\frac{\sin (60 t)}{\pi t}\right\}$

Problem 1 (c). Consider the SAME analog signal $x_{a}(t)$ with maximum frequency (bandwidth) $\omega_{M}=40 \mathrm{rads} / \mathrm{sec}$. This signal is sampled at the same rate $\omega_{s}=100 \mathrm{rads} / \mathrm{sec}$., 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.

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

For parts $(a)-(c)$, sampling above Nyquist Rate

$$
\omega_{s}=100>2 \omega_{M}=2(40)=80
$$

- So just meed to check 2 conditions on loupass filter
- is $H_{L p}(\omega)=T_{s}$ over $-40<\omega<40$ ?
- is $H_{L p}(\omega)=0$ for $(\omega)>\omega_{S} \omega_{M}=\begin{gathered}100-40 \\ =60\end{gathered}$ $=60$
(a)
 ?


$$
x_{r}(t)=x_{a}^{(t)}
$$

(b)

(c)

meets both conditions

$$
x_{r}(t)=x_{a}(t)
$$

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

$$
x[n]=x_{a}\left(n T_{s}\right)=\left\{\frac{\pi}{5} \frac{\sin \left(\frac{\pi}{8} n\right)}{\pi n T_{s}} \frac{\sin \left(\frac{3 \pi}{8} n\right)}{\pi n T_{s}}\right\} 2 j \sin \left(\frac{\pi}{2} n\right) \quad \text { where: } \quad T_{s}=\frac{2 \pi}{80}
$$

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.

$$
x_{r}(t)=\sum_{n=-\infty}^{\infty} x[n] h\left(t-n T_{s}\right) \quad \text { where: } \quad T_{s}=\frac{2 \pi}{80} \quad \text { and } \quad h(t)=T_{s} \frac{\sin (40 t)}{\pi t}
$$

Problem 1 (e). Consider the SAME analog signal $x_{a}(t)$ with maximum frequency (bandwidth) $\omega_{M}=40 \mathrm{rads} / \mathrm{sec}$. This signal is sampled at the same rate $\omega_{s}=80 \mathrm{rads} / \mathrm{sec}$., 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.

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

Problem 1 (f). Consider the SAME analog signal $x_{a}(t)$ with maximum frequency (bandwidth) $\omega_{M}=40 \mathrm{rads} / \mathrm{sec}$. This signal is sampled at the same rate $\omega_{s}=80 \mathrm{rads} / \mathrm{sec}$., 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.

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

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

Show your work for Prob. 1, parts (d)-(e)-(f).
For parts (d); $\left(e^{-}\right),(+)$, we sample rent at Nyauist Rate $\Rightarrow$ only ILPF in part (d) will work (d)

(e)


- not flat over $-40<\omega<40$
not 0 for $|\omega|>\omega_{s}-\omega_{m}=40$

$$
x_{r}(t) \neq x_{a}(t)
$$

(f)


- rot flat over $-40<\omega<40 x$
$x$ - tor $|\omega|>40$

$$
x_{r}(t) \neq x_{a}(t)
$$

Problem 1 (g). Consider an analog signal $x_{a}(t)$ with maximum frequency (bandwidth) $\omega_{M}=40 \mathrm{rads} / \mathrm{sec}$. This signal is sampled at a rate $\omega_{s}=50 \mathrm{rads} / \mathrm{sec}$., where $\omega_{s}=2 \pi / T_{s}$ such the time between samples is $\bar{T}_{s}^{-}=\frac{2 \pi}{50} \mathrm{sec}$, yielding the following discrete-time sequence:

$$
x[n]=x_{a}\left(n T_{s}\right)=\left\{\frac{\pi}{5} \frac{\sin \left(\frac{\pi}{5} n\right)}{\pi n T_{s}} \frac{\sin \left(\frac{3 \pi}{5} n\right)}{\pi n T_{s}}\right\} 2 j \sin \left(\frac{4 \pi}{5} n\right) \quad \text { where: } \quad T_{s}=\frac{2 \pi}{50}
$$

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.

$$
x_{r}(t)=\sum_{n=-\infty}^{\infty} x[n] h\left(t-n T_{s}\right) \quad \text { where: } \quad T_{s}=\frac{2 \pi}{50} \quad \text { and } \quad h(t)=T_{s} \frac{\sin (25 t)}{\pi t}
$$

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

$$
x_{\epsilon}[n]=x_{a}\left(n T_{s}+0.5 T_{s}\right)=\left\{\frac{\pi}{5}\right\}\left\{\frac{\left.\sin \left(\frac{\pi}{5}(n+0.5)\right)\right)}{\pi(n+0.5) T_{s}} \frac{\sin \left(\frac{3 \pi}{5}(n+0.5)\right)}{\pi(n+0.5) T_{s}}\right\} 2 j \sin \left(\frac{4 \pi}{5}(n+0.5)\right)
$$

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.

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

Show your work for Prob. 1, parts (g)-(h).
Part (g)
black=original Fouvier Transtorm


$$
x_{r}(t)=\frac{\pi}{5}\left\{\frac{\sin (5 t)}{\pi t}\right\}^{2} 2 j \sin (10 t)
$$

$$
\text { Part }\left(h_{1}\right)
$$

black= original Fourier Transform


$$
\begin{aligned}
x_{r}(t) & =\frac{2}{20} \frac{1}{j} \frac{d}{d t}\left\{\frac{\sin (20 t)}{\pi t}\right\} \\
& +2\left\{\frac{\sin (2.5 t)}{\pi t}\right\} 2 j \sin (22.5 t)
\end{aligned}
$$

Problem 2. Consider the input signal $x_{0}(t)$ below.

$$
x_{0}(t)=e^{-j 32 t}+e^{-j 28 t}+e^{-j 24 t}+e^{-j 16 t}+e^{-j 12 t}+e^{-j 8 t}+e^{-j 4 t}+1+e^{j 4 t}+e^{j 8 t}+e^{j 12 t}+e^{j 16 t}+e^{j 24 t}+e^{j 28 t}+e^{j 32 t}
$$

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

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

to form $x(t)=x_{0}(t) * h_{L P}(t)$, 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

$$
\begin{equation*}
h[n]=16\left\{\frac{\sin \left(\frac{\pi}{4} n\right)}{\pi n}\right\}^{2} 2 j \sin \left(\frac{\pi}{2} n\right) \tag{1}
\end{equation*}
$$

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 $h_{L P}(t)$ 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 $x[n]$.

$$
\begin{array}{rlr}
x(t)= & 0 \cdot e^{-j 32 t}+0 \cdot e^{j 32 t} \quad(k=2) \\
& -1 \cdot e^{-j 28 t}+1 \cdot e^{j 28 t} \quad(k=7) \\
& -1 e^{-j 24 t}+1 \cdot e^{j 24 t} \quad(k=6) \\
& -2 e^{-j 16 t}+2 e^{j 6 t} \quad(h=4) \\
& -2 e^{-j 12 t}+2 e^{j(2 t} \quad(k=3) \quad \begin{aligned}
& \text { substitute } \\
& t=n T_{s}
\end{aligned} \\
& \sim 3 e^{-j 8 t}+3 e^{j 8 t} \quad(k=2) \\
& -3 e^{-j 4 t}+3 e^{j 4 t} \quad(h=1)=k 4 \frac{2 \pi}{64} \\
& 0 \cdot 1 &
\end{array}
$$



## Digital Filter Frequency Response



$$
\begin{aligned}
& y[n]=(-1)(-1) e^{-j \frac{7 \pi}{8} n}+(1)(1) e^{j \frac{7 \pi}{8} n} \\
&(-2)(-1) e^{-j \frac{6 \pi}{8} n}+(2)(1) e^{j \frac{6 \pi}{8} n} \\
&\left(-\frac{1}{4}\right)(-2) e^{-j \frac{4 \pi}{8} n}+(4)(2) e^{j \frac{4 \pi}{8} n} \\
&(-3)(-2) e^{-j \frac{\pi}{8} n}+(3)(2) e^{j \frac{3 \pi}{8} n} \\
&(-2)(-3) e^{-j \frac{2 \pi}{8} n}+(2)(3)+e^{j \frac{2 \pi}{8} n} \\
&(-1)(-3) e^{-j \frac{\pi}{8} n}+(1)(3) e^{j \frac{\pi}{8} n} \\
&= 2 \cos \left(\frac{7 \pi}{8} n\right)+4 \cos \left(\frac{6 \pi}{8} n\right) \\
&+16 \cos \left(\frac{4 \pi}{8} n\right)+12 \cos \left(\frac{3 \pi}{8} n\right) \\
&+12 \cos \left(\frac{2 \pi}{8} n\right)+\cos \left(\frac{\pi}{8} n\right)
\end{aligned}
$$



