

**MAE143 A - Signals and Systems - Winter 11
Final**

Instructions

- (i) This exam is open book. You may use whatever written materials you choose, including your class notes and textbook. You may use a hand calculator with no communication capabilities
- (ii) You have 3 hours
- (iii) Do not forget to write your name, student number, and instructor

1. **Filtering and sampling.** The continuous-time signal $x(t) = 100 \text{sinc}^2(100t)$ goes through an ADC block that samples signals at a frequency $f_s = 190$ Hz. Answer the following questions

- (a) (1 point) Compute the Fourier transform of $x(t)$. What is its bandwidth?
- (b) (1 point) Would you be able to reconstruct the original signal out of the samples taken by the ADC block? Why?
- (c) (2 points) Consider a system whose impulse response is given by

$$h(t) = 150 \text{sinc}(150t)$$

What kind of filter is this? What is its cutoff frequency in Hz? Plot the magnitude and phase of the transfer function.

- (d) Suppose the signal $x(t)$ passes first through the system in (c) to produce the signal $y(t)$ and then goes through the ADC block to produce the samples, see Figure 1.

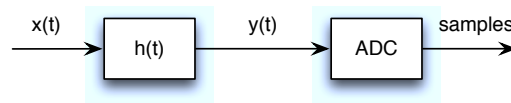


Figure 1: Block diagram for question 3, part (d).

- i. (1 point) What is the bandwidth of $y(t)$? Would you be able to reconstruct the signal $y(t)$ out of the samples taken by the ADC block? Justify your answer.
- ii. (1 point) Why would one call the system in (c) an antialiasing filter?
- (e) (1 point (bonus)) With knowledge of $y(t)$, would you be able to recover the original signal $x(t)$? Why?

Solution: (a) Using the table of basic transforms and properties (time scaling), the Fourier transform of $x(t)$ is

$$X(f) = \text{tri}\left(\frac{f}{100}\right) \quad (+.5 \text{ point})$$

The bandwidth of this signal is therefore $f_B = 100$ Hz. (+.5 point)

(b) No, because the sampling frequency is below the Nyquist rate, $f_s = 190 < 2f_B = 200$.

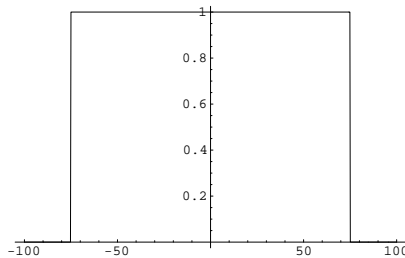
(+ 1 point)

(c) We compute the transfer function of the system as

$$H(f) = \text{rect}\left(\frac{f}{150}\right) \quad (+ 1 \text{ point})$$

Therefore, this is an ideal lowpass filter with cutoff frequency $f_c = 75$ Hz. (+ .5 point)

The phase plot of the filter is trivial (identically zero). The magnitude plot looks like



(+ .5 point)

(d.i) After the signal $x(t)$ goes through the ideal filter, all frequencies above the cutoff frequency (75 Hz) get cut. Therefore, the bandwidth of $y(t)$ is 75 Hz. (+ .5 point)

Since the sampling frequency of the ADC block is $190 > 150 = 2 * 75$, we will be able to reconstruct the signal $y(t)$ out of the samples produced by the ADC block. (+ .5 point)

(d.ii) The reason for the antialiasing name comes from the effect that this filter has on the signal. Since it makes sure that no frequency larger than $\frac{f_s}{2}$ will get into the ADC block, it eliminates the possibility of aliasing when performing the sampling. (+ 1 point)

(e) No, an ideal filter is not invertible. (+ 1 extra point)

2. **Continuous-time Fourier Series (CTFS) and system response.** Consider a periodic function described over one fundamental period ($T_0 = 4$) by

$$x(t) = \begin{cases} 1 - t, & t \in [0, 1], \\ 0, & t \in [-2, 0] \cup [1, 2]. \end{cases}$$

Do the following

(a) (.5 points) Plot $x(t)$ over the interval $[-6, 6]$. Could this signal correspond to the impulse response of a causal system? Why?

(b) (2 points) Show that the harmonic numbers of $x(t)$ with representation time $T_F = T_0$ are

$$X[0] = \frac{1}{8},$$
$$X[k] = \frac{1}{\pi^2 k^2} \left(1 - j\frac{\pi}{2}k - e^{-j\frac{\pi}{2}k} \right), \quad k \in \mathbb{Z} \setminus \{0\}.$$

(c) (1.5 points) Compute the transfer function of the LTI system

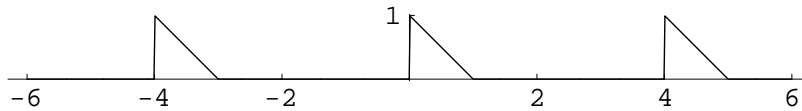
$$\dot{y}(t) + y(t) = x(t)$$

What is the impulse response? Is the system BIBO stable?

(d) (2 points) Show, using the CTFS from (b) and the transfer function from (c), that the response $y(t)$ of the LTI system to the input signal $x(t)$ is

$$y(t) = X[0] + \sum_{k=1}^{\infty} \frac{4}{\sqrt{4 + \pi^2 k^2}} |X[k]| \cos\left(\frac{\pi}{2}kt - \arctan\left(\frac{\pi k}{2}\right) + \angle X[k]\right)$$

Solution: (a) Over $[-2, 2]$, the function $x(t)$ is only nonvanishing on $[0, 1]$, where it looks like a unit triangle. It has fundamental period $T_0 = 4$. Therefore, the plot of $x(t)$ over the interval $[-6, 6]$ is



(+ .25 point)

The function cannot be the impulse response of a causal system because there are values of time before $t = 0$ for which the function is nonzero. Since the impulse is exerted at $t = 0$, a causal system cannot anticipate the future to make this happen. (+ .25 point)

(b) To compute the Fourier series of $x(t)$, we need to compute the harmonic numbers

$$X[k] = \frac{1}{T_0} \int_{t_0}^{t_0+T_0} x(t) e^{-j2\pi k f_0 t} dt \quad (+ .5 \text{ point})$$

We select $t_0 = -T_0/2 = -2$ (any choice would give the same answer since the function is periodic). For $k = 0$, we get

$$X[0] = \frac{1}{4} \int_0^1 (1-t) dt = \frac{1}{8}. \quad (+ .5 \text{ point})$$

Next, we reason for the case when $k \neq 0$. Over the interval $[-2, 2]$, the function is only non-vanishing in $[0, 1]$. Then,

$$X[k] = \frac{1}{T_0} \int_{-T_0/2}^{T_0/2} x(t) e^{-j2\pi k f_0 t} dt = \frac{1}{4} \int_0^1 (1-t) e^{-j\frac{\pi}{2}kt} dt$$

Note that, using integration by parts,

$$\int_a^b t e^{-ct} dt = -\frac{1}{c} t e^{-ct} \Big|_a^b + \frac{1}{c} \int_a^b e^{-ct} dt = -\frac{1}{c} t e^{-ct} \Big|_a^b - \frac{1}{c^2} e^{-ct} \Big|_a^b \quad (+ .5 \text{ point})$$

Substituting above, we get (for $c = j\frac{\pi}{2}k$)

$$\begin{aligned} X[k] &= \frac{1}{4} \left(-\frac{1}{c}e^{-ct} + \frac{1}{c}te^{-ct} + \frac{1}{c^2}e^{-ct} \right) \Big|_0^1 \\ &= \frac{1}{4} \left(-\frac{1}{c}e^{-c} + \frac{1}{c}e^{-c} + \frac{1}{c^2}e^{-c} \right) - \frac{1}{4} \left(-\frac{1}{c} + 0 + \frac{1}{c^2} \right) \\ &= \frac{1}{4c^2} (e^{-c} + c - 1) = \frac{1}{\pi^2 k^2} \left(1 - j\frac{\pi}{2}k - e^{-j\frac{\pi}{2}k} \right) \end{aligned} \quad (+ .5 \text{ point})$$

(c) We use the Laplace transform to get

$$sY(s) + Y(s) = X(s) \quad (+ .25 \text{ point})$$

Therefore, the transfer function is

$$H(x) = \frac{Y(s)}{X(s)} = \frac{1}{s+1} \quad (+ .25 \text{ point})$$

The impulse response of the system can be computed by calculating the inverse Laplace transform of the transfer function. Hence,

$$h(t) = \mathcal{L}^{-1}(H(s)) = e^{-t}u(t) \quad (+ .5 \text{ point})$$

The system is BIBO stable because $\int_{-\infty}^{\infty} |h(t)|dt < \infty$ (+ .5 point)

(d) Since we have the CTFS of $x(t)$ and the transfer function of the system, we know that the response is

$$y(t) = \sum_{k=-\infty}^{\infty} X[k]H\left(j\frac{\pi k}{2}\right) e^{j\frac{\pi}{2}kt} \quad (+ .5 \text{ point})$$

Let us compute the magnitude and the phase of $H\left(j\frac{\pi k}{2}\right)$. Note that

$$H\left(j\frac{\pi k}{2}\right) = \frac{1}{1 + j\frac{\pi k}{2}}$$

Therefore,

$$\left| H\left(j\frac{\pi k}{2}\right) \right| = \frac{2}{\sqrt{4 + \pi^2 k^2}}, \quad \angle H\left(j\frac{\pi k}{2}\right) = 0 - \arctan\left(\frac{\pi k}{2}\right) \quad (+ .5 \text{ point})$$

Then, we can write the response $y(t)$ as

$$\begin{aligned} y(t) &= \sum_{k=-\infty}^{\infty} \frac{2}{\sqrt{4 + \pi^2 k^2}} X[k] e^{j(\frac{\pi}{2}kt - \arctan(\frac{\pi k}{2}))} \\ &= X[0] + \sum_{k=-\infty}^{-1} \frac{2}{\sqrt{4 + \pi^2 k^2}} X[k] e^{j(\frac{\pi}{2}kt - \arctan(\frac{\pi k}{2}))} + \sum_{k=1}^{\infty} \frac{2}{\sqrt{4 + \pi^2 k^2}} X[k] e^{j(\frac{\pi}{2}kt - \arctan(\frac{\pi k}{2}))} \\ &= X[0] + \sum_{k=1}^{\infty} \frac{2}{\sqrt{4 + \pi^2 k^2}} X[-k] e^{-j(\frac{\pi}{2}kt - \arctan(\frac{\pi k}{2}))} + \sum_{k=1}^{\infty} \frac{2}{\sqrt{4 + \pi^2 k^2}} X[k] e^{j(\frac{\pi}{2}kt - \arctan(\frac{\pi k}{2}))} \end{aligned} \quad (+ .5 \text{ point})$$

Using the facts that $|X[k]| = |X[-k]|$ and $\angle X[-k] = -\angle X[k]$, we write

$$\begin{aligned}
 y(t) &= X[0] + \sum_{k=1}^{\infty} \frac{2}{\sqrt{4 + \pi^2 k^2}} |X[k]| \left(e^{-j(\frac{\pi}{2}kt - \arctan(\frac{\pi k}{2}) + \angle X[k])} + e^{j(\frac{\pi}{2}kt - \arctan(\frac{\pi k}{2}) + \angle X[k])} \right) \\
 &= X[0] + \sum_{k=1}^{\infty} \frac{4}{\sqrt{4 + \pi^2 k^2}} |X[k]| \cos \left(\frac{\pi}{2}kt - \arctan\left(\frac{\pi k}{2}\right) + \angle X[k] \right) \quad (+.5 \text{ point})
 \end{aligned}$$

3. **Continuous-time Fourier Transform (CTFT).** Consider the periodic signal

$$z(t) = \text{tri}(t - 1) * \delta_2(t)$$

Do the following

- (a) (1 point) Plot $z(t)$ over the interval $[-6, 6]$. What is the fundamental period of $z(t)$?
 (b) (2 points) Show, using only the tables of basic CTFT transforms and basic CTFT properties, that the Fourier transform of $z(t)$ is

$$Z(f) = \frac{1}{2}\delta(f) - \sum_{k=-\infty}^{\infty} \frac{2}{\pi^2(2k-1)^2} \delta\left(f - \frac{2k-1}{2}\right)$$

and sketch a plot of its magnitude $|Z(f)|$.

- (c) (3 points) Consider now the signal

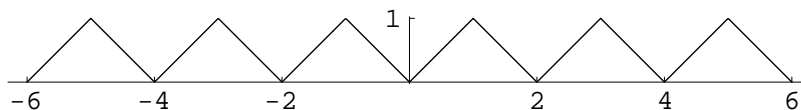
$$u(t) = z(t) \text{rect}\left(\frac{t-1}{4}\right)$$

Plot the signal $u(t)$ on the interval $[-6, 6]$. Compute the Fourier transform of $u(t)$ using the tables of basic CTFT transforms and basic CTFT properties together with the solution to (b).

- (d) (1 point (bonus)) The effect in the time domain that the multiplication by the rectangle function has on $z(t)$ is called 'windowing.' Can you explain what is the effect of windowing in the spectrum of $z(t)$?

Solution: (a) The signal $z(t)$ is a time-shifted unit triangle function with fundamental period $T_0 = 2$. (+.5 point)

Its plot looks like



(+.5 point)

(b) Since $z(t)$ is periodic, its Fourier transform corresponds to an impulse train whose coefficients are the harmonic numbers of its Fourier series. Using the multiplication-convolution property, the Fourier transform of $z(t)$ is

$$Z(f) = \mathcal{F}(\text{tri}(t - 1))\mathcal{F}(\delta_2(t)) \quad (+ .5 \text{ point})$$

From the table of transforms in the book, we know $\mathcal{F}(\text{tri}(t)) = \text{sinc}^2(f)$ and $\mathcal{F}(\delta_2(t)) = \frac{1}{2}\delta_{1/2}(f)$. Using now the time shifting property of the Fourier transform, we get

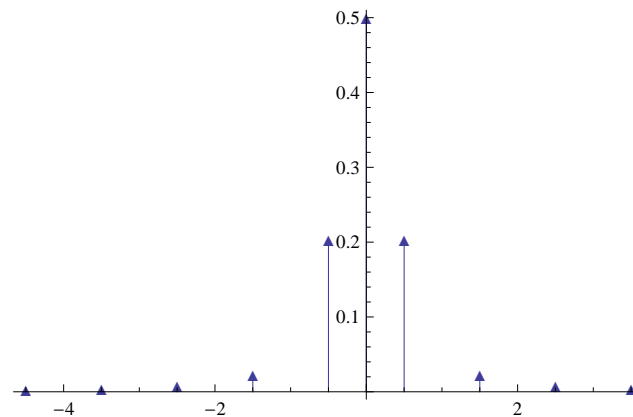
$$Z(f) = \text{sinc}^2(f)e^{-j2\pi f} \frac{1}{2}\delta_{1/2}(f) = \frac{1}{2}e^{-j2\pi f} \text{sinc}^2(f)\delta_{1/2}(f) \quad (+ .5 \text{ point})$$

$$= \frac{1}{2} \sum_{k=-\infty}^{\infty} e^{-j\pi k} \text{sinc}^2\left(\frac{k}{2}\right) \delta\left(f - \frac{k}{2}\right)$$

$$= \frac{1}{2}\delta(f) + \frac{1}{2} \sum_{k=-\infty}^{\infty} e^{-j\pi(2k-1)} \text{sinc}^2\left(\frac{2k-1}{2}\right) \delta\left(f - \frac{2k-1}{2}\right)$$

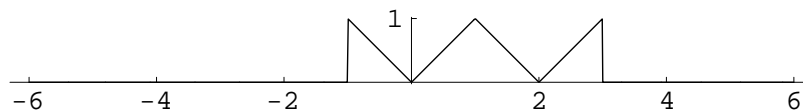
$$= \frac{1}{2}\delta(f) - \sum_{k=-\infty}^{\infty} \frac{2}{\pi^2(2k-1)^2} \delta\left(f - \frac{2k-1}{2}\right) \quad (+ .5 \text{ point})$$

The plot of $|Z(f)|$ looks like



(+ .5 point)

(c) The rectangle function $\text{rect}\left(\frac{t-1}{4}\right)$ is only nonzero over the interval $[-1, 3]$. The plot $u(t)$ over the interval $[-6, 6]$ is then



(+ .5 point)

The signal $u(t)$ is not periodic anymore. Using the multiplication-convolution property, the Fourier Transform of $u(t)$ is

$$U(f) = Z(f) * \mathcal{F}\left(\text{rect}\left(\frac{t-1}{4}\right)\right) \quad (+ .5 \text{ point})$$

From the table of transforms in the book, we know $\mathcal{F}(\text{rect}(t)) = \text{sinc}(f)$. Using the time shifting and time scaling properties of the Fourier transform, we get

$$U(f) = Z(f) * \left(4 \text{sinc}(4f)e^{-j2\pi f}\right) \quad (+ 1 \text{ point})$$

$$= \left(\frac{1}{2}\delta(f) - \sum_{k=-\infty}^{\infty} \frac{2}{\pi^2(2k-1)^2} \delta\left(f - \frac{2k-1}{2}\right)\right) * \left(4 \text{sinc}(4f)e^{-j2\pi f}\right)$$

$$= 2 \text{sinc}(4f)e^{-j2\pi f} + \sum_{k=-\infty}^{\infty} \frac{8}{\pi^2(2k-1)^2} e^{-j2\pi f} \text{sinc}(4f - 2(2k-1)) \quad (+ 1 \text{ point})$$

where we have used the fact that $e^{j\pi(2k-1)} = -1$.

(d) The effect that windowing has in the frequency domain is to spread the spectrum of the signal $z(t)$. This signal is periodic, and hence its energy is concentrated at frequencies $f = (2k-1)/2$ (deltas at these frequencies in the expression for the Fourier transform). As can be seen in our answer in (c), the Fourier transform of the signal $u(t)$ substitutes the deltas by sinc functions that spread out the energy of the original signal throughout the whole spectrum. (+ 1 extra point)