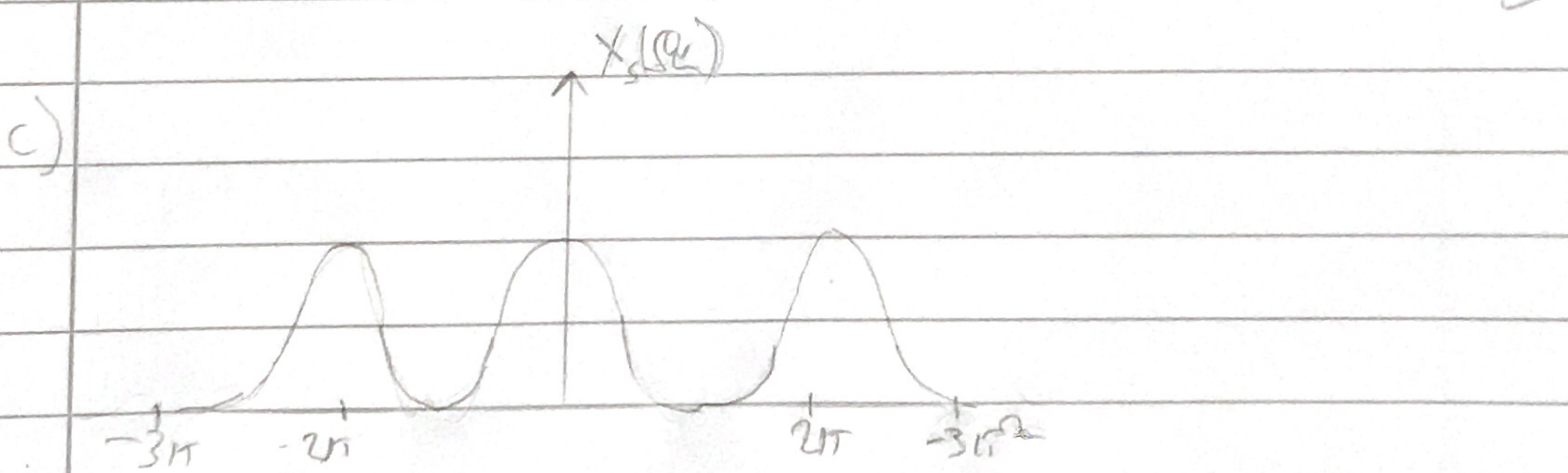


1a) No, since the Nyquist-Shannon theorem says that $w_s \geq 2w_b$ (where w_b is the signals bandwidth) is needed to avoid aliasing. And $X(w)$ have a higher bandwidth than 50 rad/s. Approximate $w_s = 120 \text{ rad/s}$ is needed to be sure that no aliasing occurs.

Generally

- b) Filter 1 is preferred since filter 2 will "cut" away the signals frequencies above 50 rad/s which will lead to that information gets lost. With filter 1 the signal is preserved, see b)*



Sampling within the frequency band yield a convolution between the original signal and the pulsetrains

$\uparrow \quad \uparrow \quad \uparrow \quad \uparrow$ This will therefore give the original signal timeshifted to all pulses.
 $-2w_s - w_s \quad w_s \quad w_s$

- *b) But if w_s is fixed chosen to be $w_s = 100 \text{ rad/s}$, aliasing will occur and distort the signal above 50 rad/s. Then it might be good to use filter 2 since this will remove the distortion from the aliasing.

29) Yes, the system is stable since all poles are within the unit circle, $|P_1| = |P_2| < 1$

b) Given by the Pole-zero map is that:

$$z_1 = -0.75 + 0.75j, z_2 = -0.75 - 0.75j$$

$$P_1 = 0.5 + 0.5j, P_2 = 0.5 - 0.5j$$

$$(z + 0.75 - 0.75j)(z + 0.75 + 0.75j) = z^2 + 1.5z + 1.125$$

$$(z - 0.5 - 0.5j)(z - 0.5 + 0.5j) = z^2 - z + 0.5$$

$$\Rightarrow H(z) = a \cdot \frac{z^2 + 1.5z + 1.125}{z^2 - z + 0.5}$$

$$H(1) = 1 \Rightarrow a \cdot \frac{1 + 1.5 + 1.125}{1 - 1 + 0.5} = 1 \Rightarrow a = \frac{1}{7.25}$$

$$\Rightarrow \boxed{H(z) = \frac{1}{7.25} \cdot \frac{z^2 + 1.5z + 1.125}{z^2 - z + 0.5} = \frac{1}{7.25} \cdot \frac{1 + 1.5z^{-1} + 1.125z^{-2}}{1 - z^{-1} + 0.5z^{-2}}}$$

$$c) H(z) = \frac{Y(z)}{X(z)} = \frac{1}{7.25} \cdot \frac{1 + 1.5z^{-1} + 1.125z^{-2}}{1 - z^{-1} + 0.5z^{-2}} \Rightarrow$$

$$Y(z)(7.25 - 7.25z^{-1} + 3.625z^{-2}) = X(z)(1 + 1.5z^{-1} + 1.125z^{-2}) \Rightarrow$$

$$7.25Y(z) - 7.25z^{-1}Y(z) + 3.625z^{-2}Y(z) = X(z) + 1.5z^{-1}X(z) + 1.125z^{-2}X(z) \Rightarrow$$

$$z^{-1}\{ \} \Rightarrow$$

$$7.25y[k] - 7.25y[k-1] + 3.625y[k-2] = x[k] + 1.5x[k-1] + 1.125x[k-2]$$

is the signals difference equation.

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2d) We know that $y[k] = x[k] * h[k] \Rightarrow Y(z) = X(z)H(z)$
Since the system is causal, we can use
the transform tables and apply Z-transform
to $x[k]$:

$$X(k) = \cos\left(\frac{\pi}{2}k\right) + \cos\left(\frac{3\pi}{4}k\right)$$

↓

$$\begin{aligned} X(z) &= \frac{1 - z^{-1} \cos\left(\frac{\pi}{2}\right)}{1 - 2z^{-1} \cos\left(\frac{\pi}{2}\right) + z^{-2}} + \frac{1 - z^{-1} \cos\left(\frac{3\pi}{4}\right)}{1 - 2z^{-1} \cos\left(\frac{3\pi}{4}\right) + z^{-2}} \\ &= \frac{1}{1 + z^{-2}} + \frac{1 + \frac{1}{\sqrt{2}}z^{-1}}{1 + \frac{2}{\sqrt{2}}z^{-1} + z^{-2}} \end{aligned}$$

\Rightarrow

$$Y(z) = H(z)X(z) = \frac{1 + 1.5z^{-1} + 1.125z^{-2}}{7.25 - 7.25z^{-1} + 3.625z^{-2}} \cdot \left(\frac{1}{1 + z^{-2}} + \frac{1 + \frac{1}{\sqrt{2}}z^{-1}}{1 + \frac{2}{\sqrt{2}}z^{-1} + z^{-2}} \right)$$

~~= 1.5z⁻¹ + times up!~~

When $Y(z)$ is derived, inverse Z-transform
would be used to obtain $y[k]$,

3 a) We know that we can extract all necessary information from the signal by using a DFT with $L=k$, further more, to obtain a spectral resolution of $\Delta F = 0.25 \text{ Hz}$ k need to be chosen as:

$$K \geq \frac{f_s}{\Delta F} = \frac{100 \text{ Hz}}{0.25 \text{ Hz}} = 400$$

therefore, 400 samples of the signal is needed to obtain the spectral resolution of $\Delta F = 0.25$.

And since the signal is time limited we know that:

$$X(\Omega) = X[l] \text{ and } X(w_i) \approx T_s X[l]$$

which can be used to approximate $X(F)$

- b) By using a rectangular window in a) we are receiving a narrow mainlobe but with strong sidelobes, i.e. more leakage. By instead using a Butter/triangle window the sidelobes would be smaller, but then the mainlobe would be wider, more smearing. By using a window with less leakage the distortion of the sidelobes would be decreased.

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3c) We know that $Y(w) = H(w) X(w)$.
Since the output $y(t)$ is also time limited we also know that $Y(\Omega_i) = X[i]$ and
 $Y(w_i) \approx T_s Y[i]$.
By then using the DFT from both $X(t)$ and
 $y(t)$ we can derive an approximation of $H(w)$
by using that $Y(w_i) \approx T_s Y[i]$ and $X(w_i) \approx T_s X[i]$