

1.

- a) B-3 is the easiest association to identify because the two pole-zero all-pass pairs determine a flat frequency response magnitude, a zero being the reciprocal-conjugate of its pole pair.

A - 2

C - 1

The difference between these two cases is that in A the poles are closer to the zeros than in C; the impact of this is that the effect of the zeros is de-emphasized for frequencies other those the zeros are aligned with ($\pm \pi/3$ rad), which creates a notch filter behavior; thus, in A-2 the gains in the frequency response magnitude for $\omega = 0$ rad and for $\omega = \pi$ rad are closer to each other than they are in C-1.

- b) All systems are stable because the poles are all inside the unit circumference; all systems have a real-valued impulse response as all poles and zeros exist as complex-conjugate pairs.
- c) Systems A and C belong to a common category of systems named "notch filters", their purpose is to reject a desired (and specific) frequency, which is achieved by placing a zero on the unit circumference according to that frequency (i.e., angle). Thus, they can be used to eliminate a sinusoidal interference having a well-defined frequency.

System B is an all-pass system, as such, its main characteristic is the group delay (as the frequency response magnitude is flat) which is the same as writing: the non-linear phase response. Thus, it can be used as a compensator of a desired group delay distortion, for example, aiming at an overall flat group-delay response (which is the same as writing: linear-phase response).

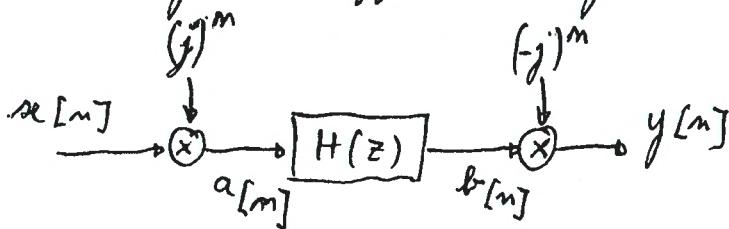
d) Given the indicated possibilities for the radii of poles and zeros, it is clear that the radii of the poles in system A is 0.25 while the radii of the poles in system B is 0.5. On the other hand, the radii of the zeros in system A is 1 while the zeros in system B is 2. Also, the orientation of poles and zeros in both systems is the same: $+\pi/3$ rad. or $-\pi/3$ rad. This means that the poles and zeros of system B are just scaled versions of the poles and zeros of system A:

$$z_B = 2 z_A \text{ . Thus: } H_B(z) = H_A\left(\frac{z}{2}\right) \text{ which, according to the properties of the Z-Transform, means that:}$$

$$h_B[n] = 2^n h_A[n] .$$

2.

a) Considering the different signals involved:



we have :

$$\begin{aligned}
 a[n] &= (j)^n x[n] \xleftarrow{Z} A(z) = X\left(\frac{z}{j}\right) \\
 b[n] &= h[n] * a[n] \xrightarrow{\quad} B(z) = H(z)A(z) \\
 &\quad = H(z)X\left(\frac{z}{j}\right) \\
 y[n] &= (-j)^n b[n] \xrightarrow{\quad} Y(z) = B\left(\frac{z}{-j}\right) \\
 &\quad = H\left(\frac{z}{-j}\right)X\left(\frac{z}{j(-j)}\right) \\
 &\quad = H(jz)X(z)
 \end{aligned}$$

Thus, $F(z) = H(jz)$

b) $H(z) = (1-z^{-1})(1+z^{-1})$ and, as $F(z) = H(jz)$, we have:

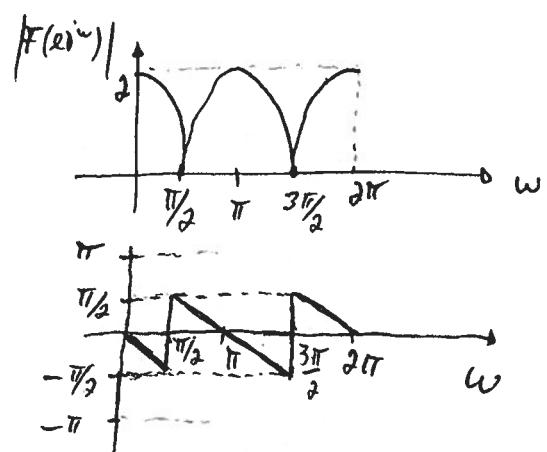
$$F(z) = (1-(jz)^{-1})(1+(jz)^{-1}) = (1+jz^{-1})(1-jz^{-1}) = 1 + z^{-2}$$

from which we obtain:

$$\begin{aligned}
 F(e^{j\omega}) &= 1 + e^{j2\omega} = e^{j\omega}(e^{j\omega} + e^{-j\omega}) = 2e^{j\omega} \cos \omega \\
 &\equiv |F(e^{j\omega})| e^{j\angle F(e^{j\omega})}
 \end{aligned}$$

where $|F(e^{j\omega})| = 2|\cos \omega|$

and $\angle F(e^{j\omega}) = -\omega +$
jumps of $\pm \pi$



c) $x_c(t) = 1 + \sin 490\pi t + \cos 480\pi t$

$$\begin{aligned}
 x[n] &= x_c(t) \Big|_{t=M T_s = \frac{m}{f_s}} = 1 + \sin 490\pi \frac{m}{280} + \cos 480\pi \frac{m}{280} \\
 &= 1 + \sin m \frac{7\pi}{4} + \cos m \frac{6\pi}{4} = e^{jm\omega_0} + \sin mw_1 + \cos mw_2
 \end{aligned}$$

which means that:

$$\omega_0 = 0 \text{ rad.}$$

$$\omega_1 = \frac{7\pi}{4} \text{ rad.}, \text{ as } |\omega_1| > \pi \therefore \omega_1 = \frac{7\pi}{4} + K2\pi = \left. \frac{7\pi + K8\pi}{4} \right|_{K=-1} = -\frac{\pi}{4} \text{ rad.}$$

$$\omega_2 = \frac{6\pi}{4} \text{ rad.}, \text{ as } |\omega_2| > \pi \therefore \omega_2 = \frac{6\pi}{4} + K2\pi = \left. \frac{6\pi + K8\pi}{4} \right|_{K=-1} = -\frac{\pi}{2} \text{ rad.}$$

which are the final frequencies within the Nyquist range, as a result:

$$\begin{aligned} x[n] &= 1 + \sin(-n\frac{\pi}{4}) + \cos(-n\frac{\pi}{2}) \\ &= 1 - \sin(n\frac{\pi}{4}) + \cos(n\frac{\pi}{2}). \end{aligned}$$

c) As the input signal components are all real-valued sinusoidal components,

$$\begin{aligned} y[n] &= |F(e^{j\theta})| e^{j(n\omega_0 + LF(e^{j\theta}))} - |F(e^{j\frac{\pi}{4}})| \sin(n\frac{\pi}{4} + LF(e^{j\frac{\pi}{4}})) \\ &\quad + |F(e^{j\frac{\pi}{2}})| \cos(n\frac{\pi}{2} + LF(e^{j\frac{\pi}{2}})) \end{aligned}$$

and considering that

$$F(e^{j\theta}) = 2$$

$$F(e^{j\frac{\pi}{4}}) = 2 e^{j\frac{\pi}{4}} \times \frac{\sqrt{2}}{2} = \sqrt{2} e^{j\frac{\pi}{4}} \therefore |F(e^{j\frac{\pi}{4}})| = \sqrt{2}, LF(e^{j\frac{\pi}{4}}) = -\frac{\pi}{4}$$

$$F(e^{j\frac{\pi}{2}}) = 2 e^{j\frac{\pi}{2}} \times 0 = 0$$

$$\text{it results that } y[n] = 2 - \sqrt{2} \sin(n\frac{\pi}{4} - \frac{\pi}{4})$$

and, presuming ideal reconstruction conditions,

$$\begin{aligned} y[n] &= y_c(t)|_{t=\frac{n}{F_s}} = 2 - \sqrt{2} \sin\left(\frac{n}{280} \frac{280\pi}{4} - \frac{\pi}{4}\right) \\ &= 2 - \sqrt{2} \sin(70\pi t - \frac{\pi}{4}), \end{aligned}$$

which leads to $y_c(t) = 2 - \sqrt{2} \sin(70\pi t - \frac{\pi}{4})$.

3.

$$a) y[n] = x[n] - x[n-1] + 0.1 y[n-1] + 0.2 y[n-2]$$

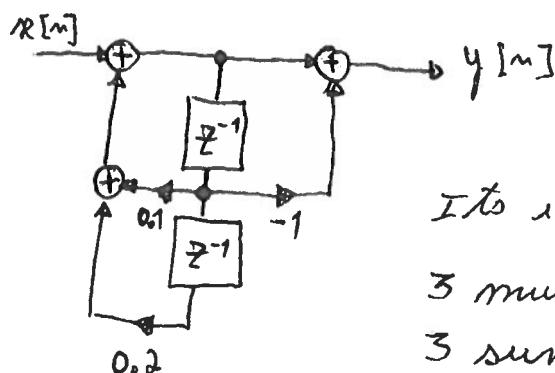
It follows directly from the difference equation that

$$Y(z) = X(z) - z^{-1}X(z) + 0.1 z^{-1}Y(z) + 0.2 z^{-2}Y(z) \quad \text{and,}$$

therefore:

$$H(z) = \frac{Y(z)}{X(z)} = \frac{1 - z^{-1}}{1 - 0.1 z^{-1} - 0.2 z^{-2}}, \text{ causal}$$

A possible realization structure is a direct type 2:



Its implementation cost is:

- 3 multiplications per output sample
- 3 sums per output sample
- 3 memory positions for coefficients
- 2 memory positions for data

b) Since $x[n] = u[n]$ then $X(z) = \frac{1}{1-z^{-1}}$, $|z| > 1$ and

$$Y(z) = H(z)X(z) = \frac{1}{1-z^{-1}} \cdot \frac{1-z^{-1}}{1-0.1z^{-1}-0.2z^{-2}} = \frac{1}{1-0.1z^{-1}-0.2z^{-2}}, \text{ causal}$$

Finding the poles:

$$z^2 - 0.1z - 0.2 = 0 \quad \therefore z = \frac{0.1 \pm \sqrt{0.01 + 0.8}}{2} = \begin{cases} 0.5 \\ -0.4 \end{cases}$$

makes it possible to write:

$$Y(z) = \frac{1}{(1-0.5z^{-1})(1+0.4z^{-1})}, |z| > 0.5$$

$$= \frac{A}{1-0.5z^{-1}} + \frac{B}{1+0.4z^{-1}}$$

where

$$A = (1-0.5z^{-1})Y(z) \Big|_{z=0.5}$$

$$= \frac{1}{1 + \frac{0.4}{0.5}} = \frac{5}{9}$$

$$B = (1+0.4z^{-1})Y(z) \Big|_{z=-0.4} = \frac{4}{9}$$

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which leads to : $Y(z) = \frac{5/9}{1-0.5z^{-1}} + \frac{4/9}{1+0.4z^{-1}}$

$$\quad \quad \quad |z| > 0.5 \quad |z| > 0.4$$

and taking the inverse Z-transform :

$$y[n] = \frac{5}{9} 0.5^n u[n] + \frac{4}{9} (-0.4)^n u[n]$$

c) Replacing \bar{z}^1 in the delay chain by \bar{z}^2 is the same as performing the same replacement in the transfer function :

$$H(z) = \frac{1-z^{-1}}{(1-0.5z^{-1})(1+0.4z^{-1})}$$

$$\xrightarrow{\bar{z}^1 \rightarrow \bar{z}^2} H(\bar{z}^2) = \frac{1-\bar{z}^2}{(1-0.5\bar{z}^2)(1+0.4\bar{z}^2)}$$

$$= \frac{(1-\bar{z}^1)(1+\bar{z}^2)}{(1-\sqrt{0.5}\bar{z}^1)(1+\sqrt{0.5}\bar{z}^1)(1-j\sqrt{0.4}\bar{z}^1)(1+j\sqrt{0.4}\bar{z}^1)}$$

Thus, the implications are :

- $|z| > \sqrt{0.5}$
- the system order increases by a factor of 2 (it becomes 4)
 - a positive zero gives rise to two zeros, one positive, another negative
 - a positive pole gives rise to two poles, one positive, and another negative; their radii is the square root of that of the initial pole
 - a negative pole gives rise to two poles, one imaginary and positive, and another imaginary and negative; their radii is the square root of that of the initial pole
 - as $H(\bar{z}^2)|_{\bar{z}=e^{j\omega}} = H(e^{j2\omega})$ then, relative to $H(e^{j\omega})$, the new frequency response consists of the initial frequency response compressed by a factor of two; this is valid for both phase and magnitude parts of the frequency response.

4.

- a) 1: $\mathbf{x} = [1 \ 0 \ 3 \ 0 \ 5 \ 0];$
- 2: $X = \text{fft}(\mathbf{x}); \ N = \text{length}(x);$
- 3: $\mathbf{Y} = \text{zeros}(\text{size}(x)); \ Y(1) = X(1);$
- 4: $Y(N:-1:2) = X(2:N);$
- 5: $Z = X.*Y; \ iifft(Z)$
- 6: $K = [0:N-1]; \ W = X.*\left(1 + \cos(K*\pi/N)\right); \ iifft(W)$

Lines 1 and 2 of the code set $\mathbf{x}[n]$ and $X[k] = \text{DFT}\{\mathbf{x}[n]\}$

lines 3 and 4 set $Y[k] = X[-k]_N = X[N-k], k=1, \dots, N-1$

Line 5 makes $Z[k] = X[k] \cdot Y[k] = X[k] \cdot X[-k]_N$

and computes $\mathbf{z}[n] = \text{IDFT}\{Z[k]\}$

Using the DFT properties :

$$\begin{array}{ccc} \mathbf{x}[n] & \xrightarrow{\text{DFT}} & X[k] \\ \mathbf{x}[-n] & \longleftarrow & X[-k] \end{array}$$

$$\mathbf{z}[n] = \mathbf{x}[n] \circledast \mathbf{x}[-n] \longleftrightarrow X[k] \cdot X[-k] = Z[k]$$

which leads to

	$k=0$	$k=N-1$	
$\mathbf{x}[k]$.. 0 1 0 3 0 5 0 1 0 ..		
$\mathbf{x}[-k]_N$	1 0 5 0 3 0		
$\mathbf{x}[-(-k)]_N = \mathbf{x}[k]$	1 0 3 0 5 0	$\sum_k \mathbf{x}[k] \mathbf{x}[k] = 35$	
$\mathbf{x}[k-1]$	0 1 0 3 0 5	$\sum_k \mathbf{x}[k] \mathbf{x}[(k-1)] = 0$	
$\mathbf{x}[k-2]$	5 0 1 0 3 0	$\sum_k \mathbf{x}[k] \mathbf{x}[(k-2)] = 23$	
$\mathbf{x}[k-3]$	0 5 0 1 0 3	$\sum_k \mathbf{x}[k] \mathbf{x}[(k-3)] = 0$	
$\mathbf{x}[k-4]$	3 0 5 0 1 0	$\sum_k \mathbf{x}[k] \mathbf{x}[(k-4)] = 23$	
$\mathbf{x}[k-5]$	0 3 0 5 0 1	$\sum_k \mathbf{x}[k] \mathbf{x}[(k-5)] = 0$	

$$\begin{aligned} \text{Thus, } \mathbf{z}[n] &= \mathbf{x}[n] \circledast \mathbf{x}[-n] = \sum_{k=0}^{N-1} \mathbf{x}[k] \mathbf{x}[-(n-k)]_N = \sum_{k=0}^{N-1} \mathbf{x}[k] \mathbf{x}[(n-k)]_N = \\ &= [35 \ 0 \ 23 \ 0 \ 23 \ 0] \end{aligned}$$

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f) Line 6 of the code sets:

$$\begin{aligned} w[k] &= x[k] \cdot (1 + \cos(k \cdot 2\pi/N)) \\ &= x[k] \cdot \left(1 + \frac{e^{j k \frac{2\pi}{N}} + e^{-j k \frac{2\pi}{N}}}{2} \right) \\ &= x[k] + \frac{1}{2} e^{j k \frac{2\pi}{N}} x[k] + \frac{1}{2} e^{-j k \frac{2\pi}{N}} x[k] \end{aligned}$$

and, given the DFT property: $\mathcal{R}[(m-m_0)_N] \longleftrightarrow e^{-j k \frac{2\pi}{N} m_0} x[k]$
 we have: $\mathcal{R}[(m+1)_N] \longleftrightarrow e^{j k \frac{2\pi}{N}} x[k]$
 $\mathcal{R}[(m-1)_N] \longleftrightarrow e^{-j k \frac{2\pi}{N}} x[k]$

which leads to:

$$w[m] = \mathcal{R}[m] + \frac{1}{2} \mathcal{R}[(m+1)_N] + \frac{1}{2} \mathcal{R}[(m-1)_N]$$

e)

as

$\mathcal{R}[m]$... 0 1 0 3 0 5 0 7 0 ...
$\mathcal{R}[(m+1)_N]$	0 3 0 5 0 1
$\mathcal{R}[(m-1)_N]$	0 1 0 3 0 5

we obtain:

$$w[m] = \boxed{\begin{bmatrix} 1 & 2 & 3 & 4 & 5 & 3 \end{bmatrix}}$$

5.

- a) As a general concept, if $x[n]$ is a discrete-time signal, its DFT, $X[k]$, is N -periodic, i.e. $X[k] = X[k+lN]$, In the discrete-frequency domain, $\forall l \in \mathbb{Z}$
 N represents the period of $X[k]$, but also represents the sampling frequency in that domain, and $\frac{N}{2}$ represents the Nyquist frequency (half of the sampling frequency).

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On the other hand, if $x[n]$ is real-valued, then, according to the DFT properties, $X[k] = X^*[N-k]$, $k=1, \dots, N-1$

$$= X^*[-k]_N$$

which means that the $X[k]$ sequence is conjugate symmetric i.e. the negative frequency axis is the conjugate mirror of the positive frequency axis. Considering the N -periodicity, this also means that when $x[n]$ is real-valued, $|X[k]|$ is mirrored with respect to the Nyquist frequency.

As in our case the sampling frequency is 8kHz , and the Nyquist frequency is 4kHz , real-valued signal components are easily identified by looking at even symmetries around 4kHz (for the same time). In this context, two real-valued signal components are identified :

- one having a constant frequency of 2kHz and existing between $t=0.1\text{s}$ and $t=0.7\text{s}$,
- another having a variable frequency around the mean frequency of 2kHz and existing between $t=0.2\text{s}$ and $t=0.6\text{s}$.

The spectrogram also reveals two complex-valued signal components (two complex sinusoids) :

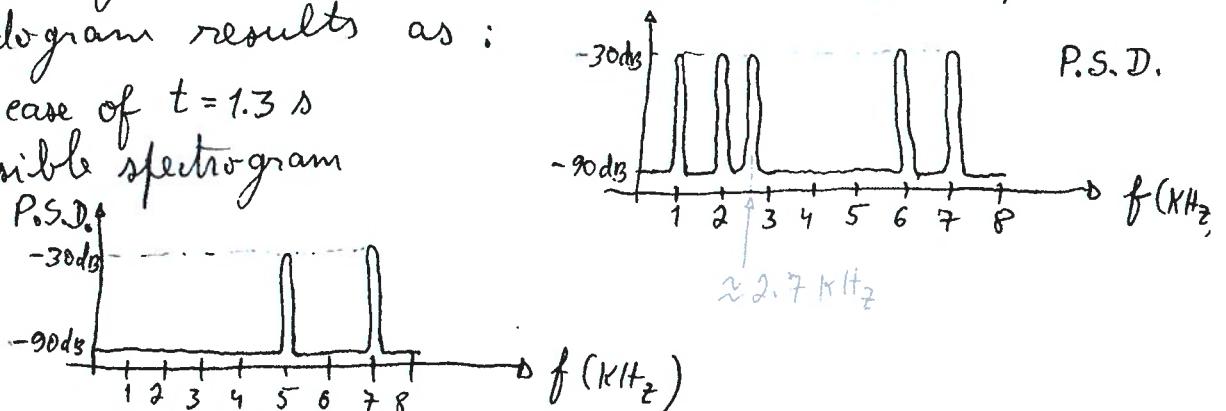
- one whose frequency increases linearly between 0Hz and the sampling frequency (8kHz) from $t=0\text{s}$ till $t=1.5\text{s}$
- another exhibiting a variable frequency around the mean frequency of 4kHz and existing between $t=1\text{s}$ and $t=1.4\text{s}$.

Spectrogram A has been obtained using the rectangular window because the fixed frequency of 2 kHz is represented by means of a thinner line than represented in spectrogram B (as a consequence of the wider main lobe width of the frequency response of the Hanning window), and because the spectrogram B is much cleaner than the spectrogram A, which is a consequence of the fact that the far-end leakage due to the Hanning window is much lower than in the case of the rectangular window.

- b) In the case of $t = 0.5 \text{ s}$, and taking a vertical line in the spectrogram (A or B, but B is easier), a plausible periodogram results as :

In the case of $t = 1.3 \text{ s}$
a plausible spectrogram

i)



NOTE : only the case $t = 0.5$ was considered for a 100% correct answer.

- c) As a result of the answer in a), we have :

- one narrow-band and real-valued signal component having a constant frequency of 2 kHz and existing between $t = 0.1$ and $t = 0.7$
- one narrow-band and real-valued signal component whose frequency varies sinusoidally around $f = 2 \text{ kHz}$, the frequency deviation is around 1 Hz, the period of the variation is $0.6 - 0.2 = 0.4 \text{ s}$, and this signal exists between $t = 0.2 \text{ s}$ and $t = 0.6 \text{ s}$.
- one narrow-band and complex-valued signal component whose center frequency increases for $t \in [0, 1.5] \text{ s}$ between 0 Hz and 8 kHz
- one narrow-band and complex-valued signal component whose center frequency is 4 kHz and varies sinusoidally for $t \in [1, 1.4] \text{ s}$, the frequency deviation is around 1 kHz and period is 0.4 s.