

Proposed Problems – DSP

P1: An analog signal

$$x_a(t) = \cos(200\pi t) - 3.5 \cos(600\pi t)$$

is sampled by 0.5 kHz. Evaluate the DFT coefficients, for $N = 20$.

P2: An analog signal $x_a(t)$ is sampled by $T = 0.01$ s. The resulting sequence is:

n	0	1	2	3	4	5
$x(n)$	5	-1.5	6.5	-3	6.5	-1.5

Evaluate the spectrum using the 6-point DFT and specify to which analog frequency corresponds each $X(k)$.

P3: Consider the analog signal

$$x_a(t) = A \cos(200\pi t) - B \cos(800\pi t)$$

sampled by 1 kHz.

- Determine the period of the obtained discrete-time signal.
- Evaluate the 20-point DFT of the sequence.
- Repeat the previous parts for a sampling frequency of 0.5 kHz. Explain the obtained result.

P4: A speech signal is sampled at a rate of 20,000 samples/sec. A segment of length 1024 samples is selected and its DFT is computed.

- What is the time duration of the segment of speech?
- What is the frequency resolution between DFT values?
- What happens if only the even samples are retained?

P5: Evaluate the 8-point DFT of the sequence $x(n) = \{1, 1, 1, 1, 1, 1, 0, 0\}$. Sketch the magnitude and the phase spectra of the obtained DFT coefficients.

P6: Evaluate the 8-point DFT for the next sequences:

$$x(n) = \cos \frac{\pi}{2} n, \quad n = \overline{0, 7};$$

$$h(n) = 2^n, \quad n = \overline{0, 7}.$$

P7: Evaluate the linear and the circular convolution ($N = 4$ and $N = 8$) of the sequences:

$$x_1(n) = \left\{ \frac{2}{\sqrt{2}}, 2, 1, 1 \right\};$$

$$x_2(n) = \left\{ 1, \frac{2}{\sqrt{2}}, 3, 4 \right\}.$$

P8: Evaluate the linear and the circular convolution ($N = 8$) of the sequences:

$$x_1(n) = \left\{ \frac{1}{\sqrt{2}}, 2, 3, 4, 5 \right\};$$

$$x_2(n) = \left\{ \frac{1}{\sqrt{2}}, -1, 2, -2, 0 \right\}.$$

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P9: Evaluate the power spectra for the periodic sequence:

$$x(n) = \left\{ \frac{2}{7}, 2, 2, 2, 0, 0, 0, 0 \right\}, \quad x(n) = x(n+8).$$

P10: A linear time-invariant system is described by the constant coefficient difference equation:

$$y(n) - \frac{3}{4}y(n-1) + \frac{1}{8}y(n-2) = 2x(n-1).$$

- Determine the homogenous solution.
- Determine the solution, if:

$$y(-1) = 1, \quad y(0) = 0 \quad \text{and} \quad x(n) = \cos \frac{2\pi n}{5} u(n).$$

P11: A linear time-invariant system is described by the constant coefficient difference equation:

$$y(n) = 0.7y(n-1) - 0.06y(n-2) + 2x(n) - x(n-2).$$

- Determine the unit impulse response.
- Determine the unit step response, in zero initial conditions.
- Evaluate the stability of the system.

P12: A linear time-invariant system is described by the constant coefficient difference equation:

$$y(n) - 5y(n-1) + 6y(n-2) = 3x(n-1).$$

- Determine the homogenous solution.
- Determine the unit impulse response.
- Determine the unit step response.

P13: Evaluate, function of $\cos \omega$, the squared magnitude of the frequency response function, for the system described by the input-output relationship:

$$y(n) = -0.1y(n-1) + 0.2y(n-2) + x(n) + x(n-1)$$

P14: Knowing that:

$$y(-1) = y(-2) = 1,$$

determine the unit step response for the system described by:

$$y(n) = 0.9y(n-1) - 0.81y(n-2) + x(n).$$

P15: A linear time-invariant system is described by the constant coefficient difference equation:

$$y(n) = 0.7y(n-1) - 0.1y(n-2) + 2x(n) - x(n-2).$$

- Determine the unit impulse response.
- Determine the unit step response.
- Evaluate the stability of the system.

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P16: Derive the radix-2 decimation-in-time FFT algorithm ($N = 16$).

P17: Determine all possible discrete time signals associated with the z -transform:

$$X(z) = \frac{5z^{-1}}{\left(1 - \frac{1}{2}z^{-1}\right)(3 - z^{-1})}.$$

For each obtained sequence specify the corresponding region of convergence.

P18:

a) Evaluate and sketch the frequency response for:

$$y(n) - \frac{1}{2}y(n-1) = x(n) + 2x(n-1) + x(n-2).$$

b) Write the constant coefficient difference equation corresponding to the system described by the frequency response function:

$$H(\omega) = \frac{1 - \frac{1}{2}e^{-j\omega} + e^{-j3\omega}}{1 + \frac{1}{2}e^{-j\omega} + \frac{3}{4}e^{-j3\omega}}.$$

P19: Consider the LTI system described by:

$$y(n) = x(n) - x(n-4).$$

a) Sketch the magnitude and the phase of the frequency response function.

b) Evaluate the output of the system to the input signal:

$$x(n) = \cos\frac{\pi}{2}n + \cos\frac{\pi}{4}n, \quad -\infty < n < \infty.$$

P20: Evaluate the unit impulse response and the frequency response, and sketch them, for the system:

$$y(n) = \frac{1}{2}y(n-1) + x(n) + \frac{1}{2}x(n-1), \quad n \in \mathbb{Z}.$$

P21: An LTI system is described by the difference equation:

$$y(n) = x(n) + x(n-10).$$

a) Sketch the magnitude and the phase of the frequency response function.

b) Evaluate the output of the system to the input signal:

$$x(n) = \cos\frac{\pi n}{10} + 3\sin\left(\frac{\pi n}{3} + \frac{\pi}{10}\right), \quad -\infty < n < \infty$$

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P22: An LTIS is described by:

$$y(n) = x(n) + 5x(n-10).$$

- Evaluate the magnitude and the phase of the frequency response function.
- Evaluate the output of the system to the input signal:

$$x(n) = 5 + 6 \cos\left(\frac{2\pi n}{5} + \frac{\pi}{2}\right), \quad -\infty < n < \infty.$$

P23: Evaluate and sketch the frequency response characteristics (magnitude, phase and group delay) of a moving average (MA) filter, for $M = 6$, with equal weights.

P24: Design a two-pole band-pass filter, with pass-band centered on $\omega = \pi/2$ and zeros in $\omega = 0$ and $\omega = \pi$. The magnitude of the frequency response function is $1/\sqrt{2}$ at $\omega = 4\pi/9$ ($\cos \frac{8\pi}{9} \cong -0.94$).

P25: Design a linear-phase FIR filter, of fourth order, for which:

$$H_r(0) = 1 \quad \text{și} \quad H_r\left(\frac{\pi}{3}\right) = \frac{1}{2}.$$

P26: Design an FIR filter which completely rejects the frequency $\omega_0 = \pi/4$. Evaluate the output of the filter to the excitation:

$$x(n) = \sin \frac{\pi n}{4} u(n), \quad n = \overline{0, 4}$$

P27: Design an FIR filter in order to reject the normalized frequencies $\frac{1}{10}$, $\frac{1}{20}$, $\frac{1}{30}$. The filter gain is 8 at $\omega = 0$. Write the transfer function as a product of second order subsystems with real-valued coefficients.

P28: Evaluate the magnitude and the phase of the frequency response function, for the system characterized by the transfer function:

$$H(z) = 1 + z^{-1} + \dots + z^{-16}.$$

For a sampling frequency of 1 kHz, determine the frequencies of the sinusoids that will be rejected by the filter.

P29: Sketch the direct forms and the lattice/lattice-ladder structure, for the LTI systems:

- $2y(n) + y(n-1) - 4y(n-3) = x(n) + 3x(n-5)$;
- $y(n) = x(n) - x(n-1) + 2x(n-2) - 3x(n-4)$.

P30: Sketch the lattice structure for the linear-phase FIR filter of length 4, for which:

$$H_r(0) = 1 \quad \text{și} \quad H_r\left(\frac{\pi}{2}\right) = \frac{1}{2}.$$

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P31: Sketch the lattice-ladder structure of the system:

$$H(z) = \frac{z^3 + z}{(z + 0.5)(z^2 + z + 0.5)}.$$

P32: Sketch the parallel and the cascade structures for the system:

$$H(z) = \frac{1 + z^{-1}}{\left(1 - \frac{1}{2}z^{-1}\right)\left(1 - 0.9e^{j\frac{\pi}{3}}z^{-1}\right)\left(1 - 0.9e^{-j\frac{\pi}{3}}z^{-1}\right)}$$

P33: Sketch the parallel and the cascade structures for the system:

$$H(z) = \frac{10\left(1 - \frac{1}{2}z^{-1}\right)\left(1 - \frac{2}{3}z^{-1}\right)(1 + 2z^{-1})}{\left(1 - \frac{3}{4}z^{-1}\right)\left(1 - \frac{1}{8}z^{-1}\right)\left[1 - \left(\frac{1}{2} + j\frac{1}{2}\right)z^{-1}\right]\left[1 - \left(\frac{1}{2} - j\frac{1}{2}\right)z^{-1}\right]}$$

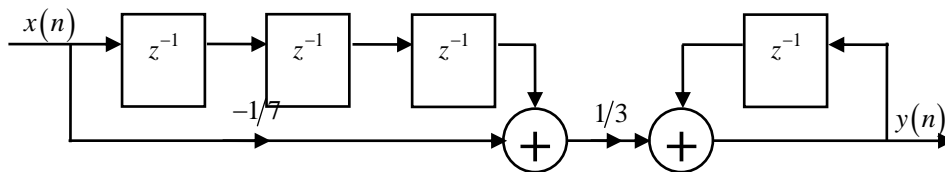
P34: Consider the system described by the transfer function:

$$H(z) = \frac{1 + \frac{1}{5}z^{-1}}{\left(1 - \frac{1}{2}z^{-1} + \frac{1}{3}z^{-2}\right)\left(1 + \frac{1}{4}z^{-1}\right)},$$

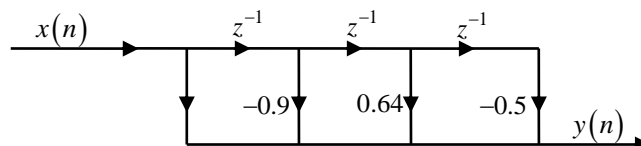
Sketch:

- a) The direct forms.
- b) The cascade structure.
- c) The parallel structure.

P35: Evaluate and sketch the unit impulse response and compute the magnitude of the frequency response function for the system described by the block diagram:



P36: Realize the lattice structure for the FIR described by the block diagram:



P37: Realize the lattice structure and evaluate the response to the unit ramp sequence, in zero initial conditions, for the system described by the transfer function:

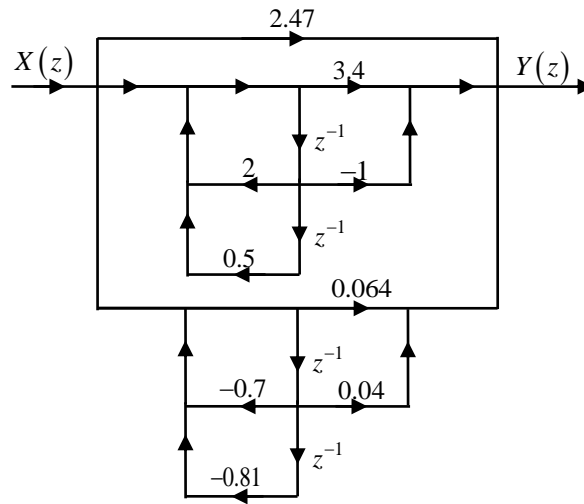
$$H(z) = \frac{1}{1 + 2z^{-1} + \frac{1}{3}z^{-2}}$$

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P38: Determine the unit impulse response of the FIR filter with lattice parameters $k_1 = 0.6$, $k_2 = 0.3$, $k_3 = 0.5$ and $k_4 = 0.9$. Evaluate the frequency response function.

P39: Determine all possible FIR filters with lattice parameters $k_1 = 1/2$, $k_2 = 0.6$, $k_3 = -0.7$ and $k_4 = 1/3$. Compute their responses to the unit step sequence.

P40: Sketch the lattice-ladder structure for the system described by the block diagram:



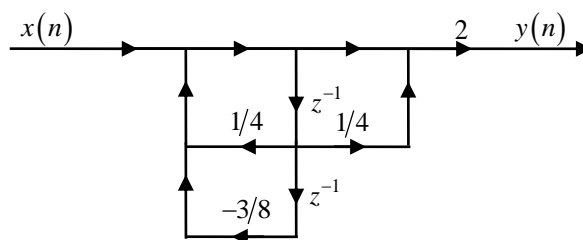
P41: Evaluate the magnitude, the phase of the frequency response function and the group delay, for the linear time-invariant system:

$$y(n) = 1.8y(n-1) - 0.81y(n-2) + x(n) + 0.95x(n-1)$$

P42: Consider the FIR filter with the lattice coefficients $k_1 = 0.65$, $k_2 = -0.34$ and $k_3 = 0.8$.

- a) Evaluate the impulse response.
- b) Sketch the direct form.

P43: Consider the system described by the block diagram:



- a) Evaluate the transfer function.
- b) Determine the impulse response.
- c) Write the input-output relationship.

P44: Realize the lattice structure and evaluate the response to the unit ramp sequence, in zero initial conditions, for the system described by the transfer function:

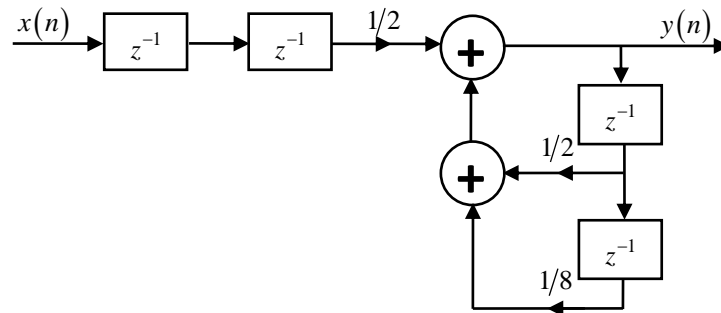
$$H(z) = 1 + 2z^{-1} + \frac{1}{3}z^{-2}.$$

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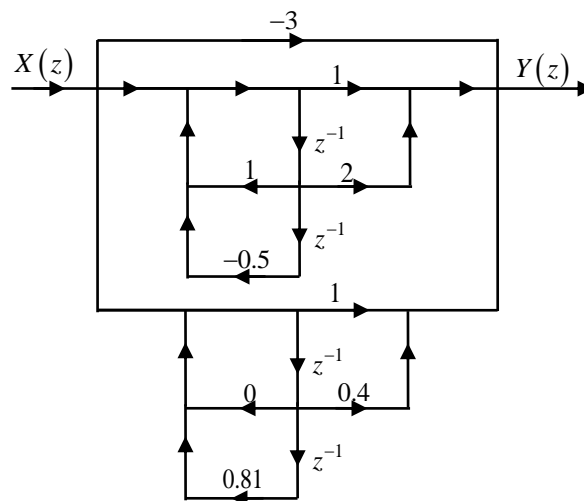
P45: Sketch the parallel and the cascade structures for the system:

$$H(z) = \frac{3 + 5z^{-1} - 2z^{-2}}{\left(1 - \frac{1}{2}z^{-1}\right)\left(1 - 3z^{-1}\right)\left(1 - \frac{1}{4}z^{-1}\right)}$$

P46: Evaluate the unit impulse response and the frequency characteristics (magnitude and phase) for the system described by the block diagram:



P47: Consider the system described by the block diagram:



Evaluate:

- Evaluate the transfer function.
- Evaluate the frequency response function.
- Determine the impulse response.
- Write the input-output relationship.

P48: Convert the analog filter with system function

$$H_a(s) = \frac{s + 0.1}{(s + 0.1)^2 + 9}$$

into a digital IIR filter by means of the impulse invariance method ($T = 0.1$ s and $T = 0.5$ s). For which value of T , the alias effect is more prevalent?

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P49: Compute the Fourier transform for the following sequences:

$$\begin{aligned} \text{a) } x(n) &= \begin{cases} 1, & n = \overline{0, M} \\ 0, & \text{otherwise} \end{cases} \\ \text{b) } x(n) &= \begin{cases} \frac{1}{2} \left(1 + \cos \frac{2\pi n}{M} \right), & n = \overline{0, M} \\ 0, & \text{otherwise} \end{cases} \end{aligned}$$

P50: Evaluate the unit impulse response of the system:



Proof that:

$$h(n) = [\delta(n) + \delta(n - 1)] * \left[\left(\frac{1}{2} \right)^n u(n) \right]$$

where ‘*’ stands for ‘linear convolution’.