

## Exercises for Digital Signal Processing

1.

Determine whether each of the discrete-time signals listed as follows is periodic or not. If it is periodic, find its period.

(a)  $\frac{6}{5} \cos\left(\frac{72}{73}n\right)$

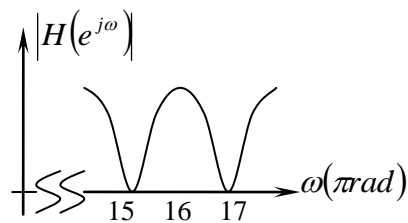
(b)  $\frac{1}{5} e^{j\left(\frac{\pi}{8}\right)} + 1$

2.

The unit impulse response of a LTI system  $h[n]$  has a length of 9, and the input of the system  $x[n]$  has a length of 128. Then the zero-state response of the system should have a length of \_\_\_\_\_.

3.

A gain curve of a digital filter is shown in Figure 1. What type could it be, LPF, HPF, BPF, or BSF? \_\_\_\_\_



**Figure 1**

4.

Consider a discrete-time LTI system with transfer function:

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

- (a) Specify the poles and zeros of the system, and then sketch the pole-zero plots for the system on the z-plane.
- (b) If the system is stable, determine the unit impulse response  $h[n]$  and the frequency response of the system.
- (c) What is the unit impulse response  $h[n]$  of the system if it is causal?

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(d) Sketch the diagram of direct form 2 realization for this system.

5.

Suppose  $x(t)$  is a continuous-time signal with the continuous-time Fourier transform  $X(j\Omega)$  shown in Figure 2. We want to get a discrete-time sequence  $x_4[n]$  from the system shown in Figure 3, where the gain of the Ideal Discrete-time LPF is 1 throughout the passband.

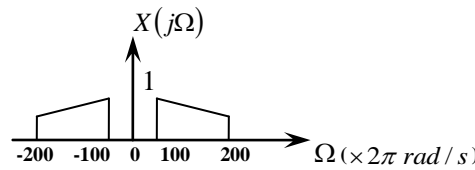


Figure 2

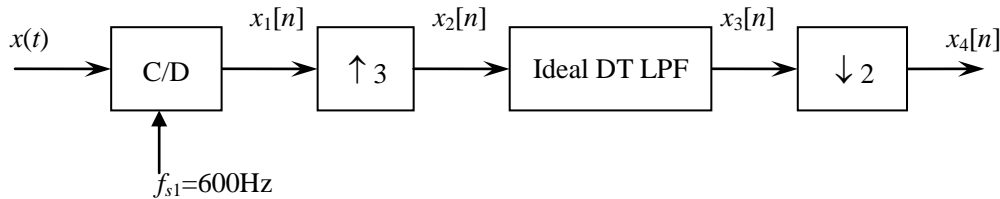


Figure 3

- (a) The output signal  $x_4[n]$  should be the sampling version of  $x(t)$  at a new sampling frequency  $f_{s2}$ . Determine the value of  $f_{s2}$ .
- (b) To avoid aliasing in f-domain and reverse mirror-image, determine the cut-off frequency of the ideal discrete-time LPF.
- (c) Sketch the spectra of each sequence  $x_1[n]$ ,  $x_2[n]$ ,  $x_3[n]$  and  $x_4[n]$ .

6.

A LTI system has a unit impulse response  $h[n] = R_6[n]$ , and the input of the system is  $x[n] = 3^n[u[n] - u[n-3]]$ . Let  $X(k)$  and  $H(k)$  be the  $N$ -point DFT of  $x[n]$  and  $h[n]$  respectively, and given

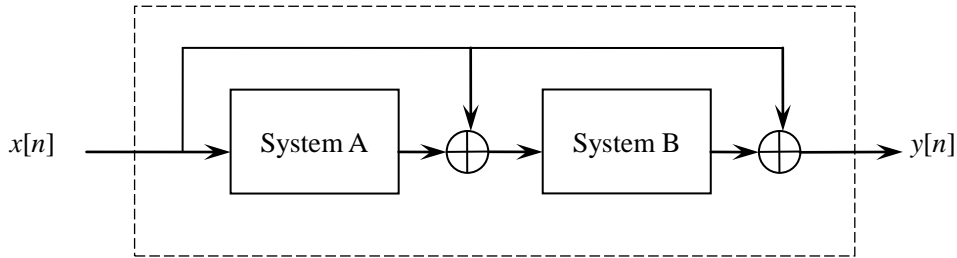
$$y_c[n] = \text{IDFT}_N[X(k)H(k)], \quad n, k = 0, 1, 2, 3, \dots, N-1$$

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- (a) Find the zero-state response of the system  $y_l[n]$ . Sketch and label it carefully.
- (b) If  $N=6$ , determine  $y_c[n]$ , then sketch and label it carefully.
- (c) If  $N=8$ , determine  $y_c[n]$ , then sketch and label it carefully.

7.

Two linear time-invariant discrete-time systems are connected as shown in Figure 4. System A has a unit impulse response  $h_1[n] = \delta[n] + 2\delta[n-1]$  and System B has a unit impulse response  $h_2[n] = u[n] - u[n-3]$ .



**Figure 4**

- (d) Are both System A and System B causal? Justify your answer.
- (e) Determine the unit impulse response of the overall system.
- (f) Is the overall system stable? Justify your answer.

8.

Consider these transfer functions for FIR filters shown as follows:

- (i)  $H_1(z) = 1 + 0.87z^{-1} + 1.1z^{-2} - 1.1z^{-4} - 0.87z^{-5} - z^{-6}$
- (ii)  $H_2(z) = 1 + 0.707z^{-2} + 0.54z^{-3} - 0.707z^{-4} - z^{-6}$
- (iii)  $H_3(z) = 1 + z^{-7}$

- (a) Determine whether each filter has a linear phase characteristic or not, justify your answer.
- (b) If the filter has a linear phase characteristic, find its phase function  $\theta(\omega)$ .

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9.

The transfer function of a 2nd order analog Butterworth LPF is:

$$H_a(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$$

Let the sampling period  $T=0.5s$ , the cutoff frequency of the filter be  $1\text{rad/sec}$ , transform this analog filter to a digital filter by bilinear transformation method.

- (a) Find the transfer function  $H(z)$  for the digital LPF.
- (b) Calculate the 3dB cutoff frequency for the digital LPF.

10.

Using windows method to design a FIR LPF with linear phase. Specifications are given as follows:

The pass band cut-off frequency  $\omega_p = 0.2\pi$  rad,

The stop band cut-off frequency  $\omega_s = 0.35\pi$  rad,

The minimum attenuation in stopband  $\alpha_s = 60\text{dB}$ ,

The maximum attenuation in passband  $\alpha_p = 3\text{dB}$ ,

- (a) Find the transition width and cut-off frequency.
- (b) Determine the frequency response  $H_d(e^{j\omega})$  and impulse response  $h_d[n]$  for the ideal filter.
- (c) Choose the window type and length according to Table 1.
- (d) Find the unit impulse response  $h[n]$  for the FIR filter.

Table 1 Window function guidelines

Window Type	Window Function $0 \leq n \leq M$	Approximate Transition width	Peak Approximation Error, $20\log_{10} \delta$ (dB)
Rectangular	1	$\frac{1.8\pi}{M}$	-21
Bartlett	$\begin{cases} \frac{2n}{M}, & 0 \leq n \leq \frac{M}{2} \\ 2 - \frac{2n}{M}, & \frac{M}{2} \leq n \leq M \end{cases}$	$\frac{6.1\pi}{M}$	-25
Hanning	$0.5 - 0.5\cos\left(\frac{2\pi n}{M}\right)$	$\frac{6.2\pi}{M}$	-44
Hamming	$0.54 - 0.46\cos\left(\frac{2\pi n}{M}\right)$	$\frac{6.6\pi}{M}$	-53
Blackman	$0.42 - 0.5\cos\left(\frac{2\pi n}{M}\right) + 0.08\cos\left(\frac{4\pi n}{M}\right)$	$\frac{11\pi}{M}$	-74