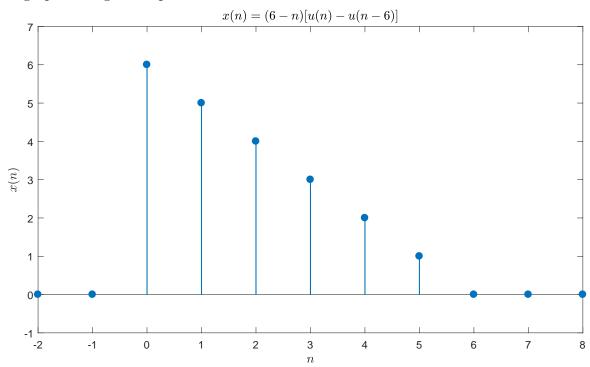
# Midterm: April, 2017

# Solved by Le Diep Phi

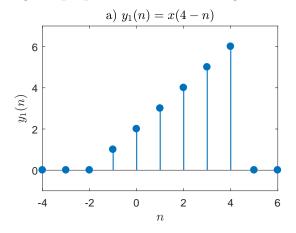
November 4, 2020

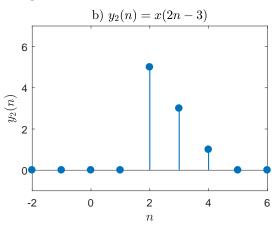
# Question 1

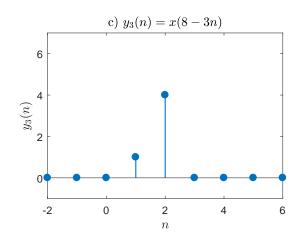
The graph of original sequence

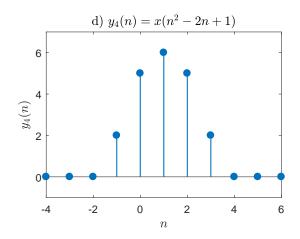


Using the properties of time shifting and time scaling, we obtain the results as follows:







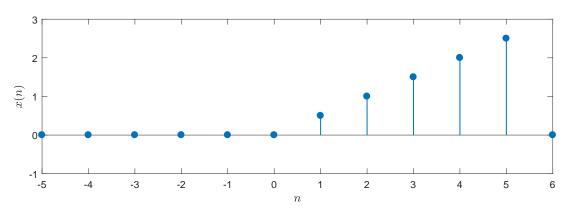


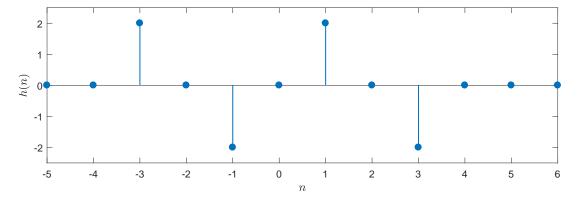
# Question 2

a) Value table for x(n) and h(n)

								n									
x(n)	0	0.5	1.0	1.5	2.0	2.5	0	h(n)	0	2	0	-2	0	2	0	-2	0

The graph for x(n) and h(n) are sketched as follows





b) From the graph of h(n), we can express h(n) as sum of impulse train, that is,

$$h(n) = 2\delta(n+3) - 2\delta(n+1) + 2\delta(n-1) - 2\delta(n-3)$$

By the definition of impulse response, we obtain the I/O relation ship via the impulse response above

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

Therefore, the output of the system is

$$+ y(-3) = 2x(0) - 2x(-2) + 2x(-4) - 2x(-6) = \dots = 0$$

$$+ y(-2) = 2x(1) - 2x(-1) + 2x(-3) - 2x(-5) = \dots = 1$$

$$+ y(-1) = 2x(2) - 2x(0) + 2x(-2) - 2x(-4) = \dots = 2$$

$$+ y(0) = 2x(3) - 2x(1) + 2x(-1) - 2x(-3) = \dots = 2$$

$$+ y(1) = 2x(4) - 2x(2) + 2x(0) - 2x(-2) = \dots = 2$$

$$+ y(2) = 2x(5) - 2x(3) + 2x(1) - 2x(-1) = \dots = 3$$

$$+ y(3) = 2x(6) - 2x(4) + 2x(2) - 2x(0) = \dots = -2$$

$$+ y(4) = 2x(7) - 2x(5) + 2x(3) - 2x(1) = \dots = -3$$

$$+ y(5) = 2x(8) - 2x(6) + 2x(4) - 2x(2) = \dots = 2$$

$$+ y(6) = 2x(9) - 2x(7) + 2x(5) - 2x(3) = \dots = 2$$

$$+ y(7) = 2x(10) - 2x(8) + 2x(6) - 2x(4) = \dots = -4$$

$$+ y(8) = 2x(11) - 2x(9) + 2x(7) - 2x(5) = \dots = -5$$

Thus,

$$y(n) = \{0, 1, 2, 2, 2, 3, -2, -3, 2, 2, -4, -5\}$$

## Question 3

**Hint:** The output for sinusoidal input signal frequency  $\omega_0$ , that is,  $x(n) = \sin(\omega_0 n + \theta)$  interacts with system by transfer function  $H(e^{j\omega})$  is given by

$$y(n) = |H(e^{j\omega_0})| \sin(\omega_0 n + \theta + \angle H(e^{j\omega_0}))$$

This equality also holds for cosine signal by replace the term sine by cosine.

#### Digital Signal Processing

### Question 3 (cont)

Sampling the analog input signal  $x_a(t)$  at frequency  $f_s = 2$  kHz we obtain the discrete time signal x(n) as below

$$x(n) = x_a(nT_s) = \sin\left(1000\pi n \times \frac{1}{2000}\right) = \sin(0.5\pi n)$$

The sampled signal has digital frequency that is  $\omega_0 = 0.5\pi$ . The magnitude and phase of the discrete time filter at that frequency is

$$H(e^{j0.5\pi}) = 2 \angle 0.9273$$

Therefore, the response y(n) of the filter after x(n) passed is given by

$$y(n) = 2\sin(0.5\pi n + 0.9273)$$

Assume that the analog re-constructor is ideal, we obtain the analog reconstructed signal  $y_a(t)$  by substituting back  $n = t/T_s$ 

$$y_a(t) = 2\sin(1000\pi t + 0.9273)$$

### Question 4

a) Assume that the quantization error e as a random variable which is distributed uniformly over the range [-Q/2, Q/2] then having probability density

$$p(e) = \begin{cases} \frac{1}{Q} & \text{if } -\frac{Q}{2} \le e < \frac{Q}{2} \\ 0 & \text{otherwise} \end{cases}$$

The mean error is

$$\bar{e} = \int_{-\infty}^{+\infty} ep(e)de = \int_{-Q/2}^{Q/2} e \frac{1}{Q} de = 0$$

The second moment error is

$$e^{-2} = \int_{-\infty}^{+\infty} e^2 p(e) de = \int_{-Q/2}^{Q/2} e^2 \frac{1}{Q} de = \frac{Q^2}{12}$$

Therefore, the average noise power is

$$\sigma_e^2 = \bar{e^2} - (\bar{e})^2 = \frac{Q^2}{12}$$

b) The non-normalized signal-to-noise ratio is given by

$$SQNR = 10 \log \left(\frac{\sigma_x^2}{\sigma_e^2}\right) = 10 \log \left(\frac{\sigma_x^2}{\frac{Q^2}{12}}\right) = 10 \log \left(\frac{12\sigma_x^2}{Q^2}\right)$$

The quantization width Q of the quantizer is given by

$$Q = \frac{R}{2^{B+1}} = \frac{2X_{\text{max}}}{2^{B+1}}$$

Then the SQNR becomes

$$\mathrm{SQNR} = 10 \log \left( \frac{12\sigma_x^2}{\frac{2^2 X_{\mathrm{max}}^2}{4^{B+1}}} \right) = 10 \log \left( \frac{3 \cdot 4^{B+1} \sigma_x^2}{X_{\mathrm{max}}^2} \right)$$

Expand and simplify this equality, we obtain

SQNR = 
$$6.02(B+1) + 4.77 - 20 \log \left(\frac{X_{\text{max}}}{\sigma_x}\right)$$

c) The problem gives us

$$SQNR = 6.02(B+1) + 4.77 - 20 \log \left(\frac{X_{max}}{\sigma_x}\right) \ge 90 \text{ dB}$$

With  $X_{\text{max}} = 3\sigma_x$ , solving for B yields that

Rounding B, we get B=15 bits. Thus, the A/D converter needs B+1=16 bits to ensure that SQNR at least 90 dB.

## Question 7

a) Observing that the impulse function at the input can express as

$$\delta(n) = \frac{1}{2}x_1(n) - \frac{1}{2}x_2(n) + x_3(n)$$

It is known that the system is linear then the above result holds that

$$L\{\delta(n)\} = \frac{1}{2}y_1(n) - \frac{1}{2}y_2(n) + y_3(n) \quad (1)$$

#### Digital Signal Processing

Again, observing the operation of the system gives us

$$\delta(n-1) = \frac{1}{2}x_2(n) - \frac{1}{2}x_x(n)$$

It is known that the system is linear then the above result holds that

$$L\{\delta(n-1)\} = \frac{1}{2}y_2(n) - \frac{1}{2}y_1(n) \quad (2)$$

From (1) and (2), it leads to the conclusion that the given system is **not time invariant**.

b) Since, we have

$$h(n) = L\{\delta(n)\} = \frac{1}{2}y_1(n) - \frac{1}{2}y_2(n) + y_3(n)$$

And

$$y_1(n) = -\delta(n+1) + 3\delta(n) + 3\delta(n-1) + \delta(n-3)$$

$$y_2(n) = -\delta(n+1) + \delta(n) - 3\delta(n-1) - \delta(n-3)$$

$$y_3(n) = 2\delta(n+2) + \delta(n+1) - 3\delta(n) + 2\delta(n-2)$$

Therefore, the impulse response of the system can be

$$h(n) = 2\delta(n+2) + \delta(n+1) - 2\delta(n) + 3\delta(n-1) + 2\delta(n-2) + \delta(n-3)$$

