

SEL4223 Digital Signal Processing

Discrete-Time Signal

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Introduction

- Digital Signal Processing (DSP) has developed rapidly over the past 40 years due to significant advances in digital computer technology and integrated-circuit fabrication (MSI→LSI→VLSI).
- In telecommunication:
 - ADC samples an analog signal and encodes the signal into pulse code modulation (PCM) format.
 - Filter noise and interference from signal, transcode, compress, and encrypt the signal prior to placing it on a carrier of an electrically based digital circuit for transmission.
 - Sending the signal to DAC for decoding back into real world analog form, perhaps with improved clarity, a shifted frequency, or demodulated for transmission.







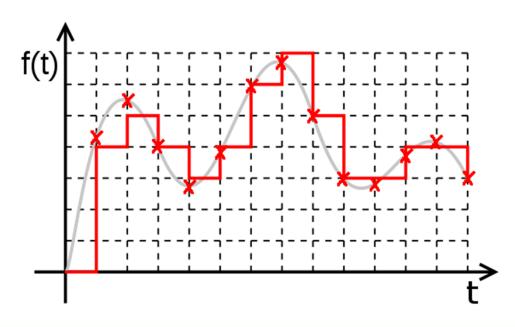
- **Signal** is defined as any physical quantity that varies with time, space or any other independent variable or variables.
- A discrete-time signal is a series consisting of a sequence of quantities. In other words, it is a series that is a function over a domain of discrete integers. Each value in the sequence is called a sample.
- As most of the signals encountered in science and engineering are analog in nature, a process on converting the analog signal to discrete signal is needed. This process is called digitization, which consists of Sampling and Quantization process.





Digitization

- Referring to figure below,
 - converting continuous value to discrete value on horizontal-axis is called sampling
 - converting the continuous value to discrete value on verticalaxis is called quantization.







Digitization (cont.)

- In the previous figure, the continuous signal is first sampled where values of f(t) at only selected t are kept. This is shown by the red cross. So far only values on the horizontal have been digitized.
- To digitize the vertical axis, values on the red cross are quantized where in this example these values are round toward nearest values of 3 bit integer. For example the first red cross value is quantized from 4.2 to 4 (or 100 in binary).





Classification of Signal

- Continuous-time and discrete time Continuous time signal take on values in the continuous interval while discrete time signal take samples only at certain values of time.
- Periodic and aperiodic Periodic signal is consistently repeated at certain time period from -∞ to ∞.
- Deterministic and random Deterministic signal can be represented with mathematical description while random signal is not where the signal evolves in time in an unpredictable manner.





• The sampling process can be viewed as

$$t = nT_s$$

- t -continuous time (s)
- n discrete time (integer). Also called sample
- T_s sampling time (s)





Sampling Theorem

• To ensure the digitized discrete-time signal represents its original continuous-time signal sufficiently and correctly, the sampling process need to follow Nyquist theorem as below:

 $F_s \ge 2F_N$

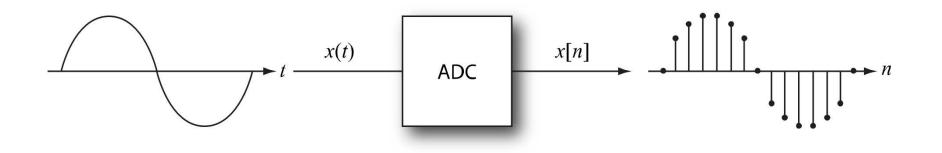
- Where:
 - F_s Frquency sampling
 - F_N Highest frequency component in the signal





Signal representation

- x(t) analog/continuous
- x[n] digital/discrete





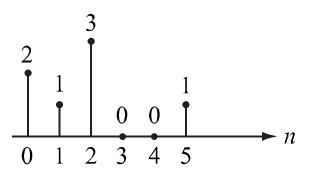


Discrete Signal Representation

Numerical method:

- x[n] = [2,1,3,0,0,1], where the arrow shows the position at n = 0.
- If no arrow showed, the first sample holds the position at n = 0

Graphical method:



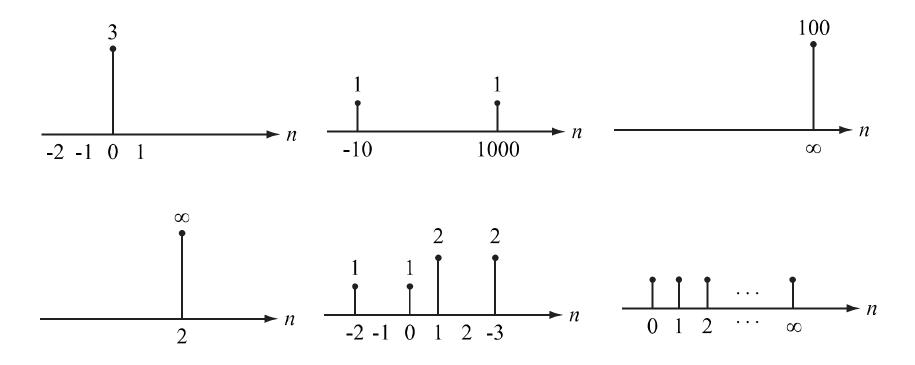
• Discrete signal length (also called size): Number of samples in the signal. Count from the lowest *n* to the highest *n* that has a nonzero value





Quiz 1

1. What is the length of these signals?



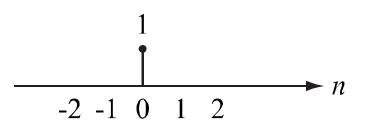
2. What is the size of a continuous signal?



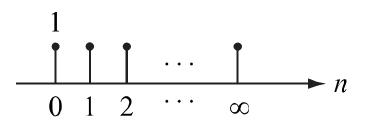


Types of Discrete Signal

- There are 2 basic types of discrete signals: impulse signal and step signal
- $\delta[n]$ impulse signal



• u[n] - step signal

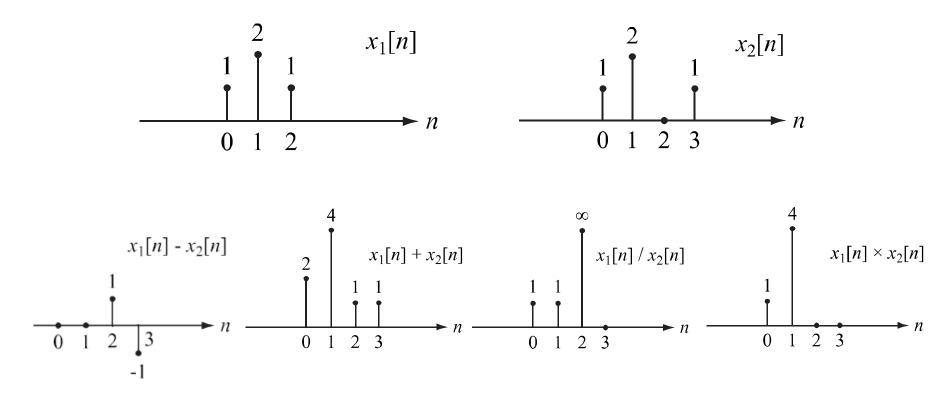






Arithmetic Operations

• For addition, subtraction, multiplication, division operations, the operations are made on each of the sample of similar *n*

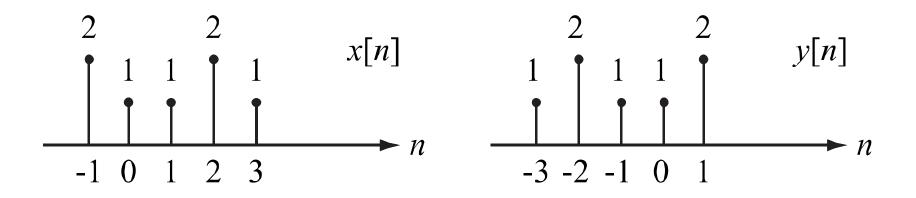






Folding Operation

• Fold signal at n = 0 where y[n] = x[-n]

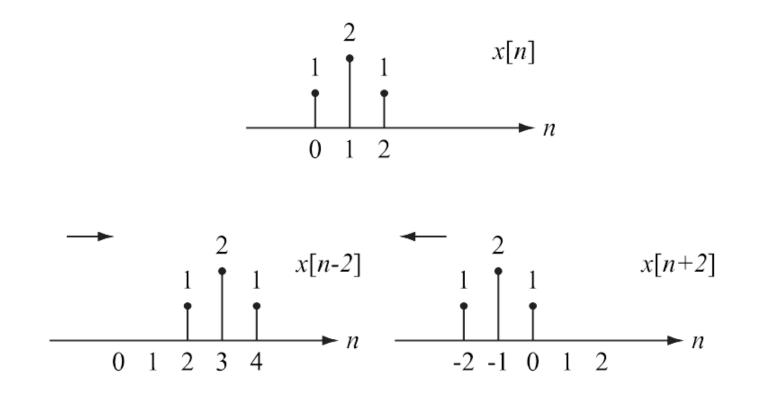






Delay Operation

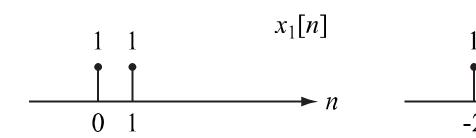
• Shift the signal to the right or left by n_d samples $y[n] = x[n - n_d]$

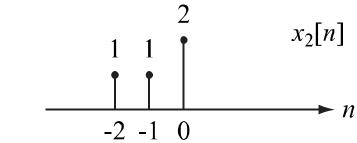












Find:

- 1) $y[n] = x_1[-n] + x_2[n]$
- 2) $y[n] = x_1[n] + x_2[n-3]$
- 3) $y[n] = x_2[-n-2]$
- 4) $y[n] = x_1[n] + x_2[-2n]$
- 5) $y[n] = x_2[n] x_1[n]$

- 6) $y[n] = u[-n] + x_1[n]$
- 7) $y[n] = \delta[n-3] + x_2[-n] + x_1[n+2]$
- 8) $y[n] = \delta[n] \times x_2[n]$
- 9) $y[n] = u[n] \times x_2[n]$
- $10)y[n] = u[-n+2] x_1[n]$
- $11) y[n] = x_2[n] + x_2[n-1]$





References

- John G. Proakis, Dimitris K Manolakis, "Digital Signal Processing: Principle, Algorithm and Applications", Prentice-Hall, 4th edition (2006).
- 2) Sanjit K. Mitra, "Digital Signal Processing-A Computer Based Approach", McGraw-Hill Companies, 3rd edition (2005).
- 3) Alan V. Oppenheim, Ronald W. Schafer, "Discrete-Time Signal Processing", Prentice-Hall, 3rd edition (2009).