

## CSE\_223\_1.pdf: Signals, Systems, and Signal Processing

### Introduction to Signals

- A **signal** is defined as any physical quantity that varies with time, space, or any other independent variable.
- Mathematically, a signal is a function of one or more independent variables:
  - Examples include sound waves, vibrations of a spring, tsunami waves, surface ripples on water, seismic waves, and electromagnetic waves.
  - An image is an example of a signal with two independent variables.
- **Signal Generation** is associated with a system responding to a stimulus or force. For example, in speech, the system includes the vocal cords and the vocal tract.

### Systems and Signal Processing

- A **system** is a device or software that performs an operation on a signal.
  - For instance, a filter reducing noise is an example of a system.
- **Signal processing** is the operation performed by a system on a signal, which is characterized by the type of operation the system performs.
  - For example, a filter eliminates noise.
- Signal processing can be done using:
  - **Analog signal processing systems.**
  - **Digital signal processing systems.**
- Most signals encountered in science and engineering are **analog** in nature. Analog signals are functions of a continuous variable (time or space) and take on values in a continuous range. These can be processed directly by analog systems.
- **Digital signal processing systems** perform processing digitally and may use a large programmable computer, a small microprocessor, or a hardwired digital processor.

### Advantages of Digital over Analog Signal Processing

- **Flexibility:** Digital processing allows easy reconfiguration by changing the program, while analog requires hardware redesign.
- **Accuracy:** Digital systems provide better control of accuracy requirements, while analog components have tolerances.
- **Storage and Transportability:** Digital signals can be stored and processed offline, which is not possible with analog signals.
- **Sophisticated Algorithms:** Digital systems allow the implementation of more sophisticated signal processing algorithms.

## Classification of Signals

- **Multichannel Signals:** Signals generated by multiple sources or sensors (e.g., ground acceleration due to an earthquake).
- **Multidimensional Signals:**
  - A signal that is a function of a single independent variable is called a **one-dimensional signal**.
  - A signal is called an **M-dimensional signal** if its value is a function of  $M$  independent variables.
  - A black-and-white TV picture can be represented as  $I(x, y, t)$ , a three-dimensional signal.
  - A color TV picture is a three-channel, three-dimensional signal with intensity functions  $I_r(x, y, t)$ ,  $I_g(x, y, t)$ , and  $I_b(x, y, t)$ .
- **Continuous-Time Signals:** Defined for every value of time and take on values in a continuous interval.
- **Discrete-Time Signals:** Defined only at specific values of time, which are usually equidistant. They can arise by sampling an analog signal or by accumulating a variable over time.
- **Continuous-Valued Signals:** Can take on all possible values in a finite or infinite range.
- **Discrete-Valued Signals:** Can take on values from a finite set of possible values.
- An **analog signal** is a continuous-time, continuous-valued signal.
- A **digital signal** is a discrete-time signal having a set of discrete values.
- **Deterministic Signals:** Can be uniquely described by an explicit mathematical expression, a table of data, or a well-defined rule. All past, present, and future values are known precisely without any uncertainty.
- **Random Signals:** Cannot be accurately described by explicit mathematical formulas and evolve unpredictably.

## Concept of Frequency in Continuous-Time and Discrete-Time Signals

- The nature of time (continuous or discrete) affects the nature of frequency.
- **Continuous-Time Sinusoidal Signals:**
  - A simple harmonic oscillation can be described as:

$$S(t) = A \sin(\omega t + \theta)$$

- For every fixed frequency  $F$ ,  $x_a(t)$  is periodic.
- Distinct frequencies result in distinct signals.

- Increasing the frequency  $F$  increases the rate of oscillation.
- Mathematically, negative frequencies are introduced for convenience.

- **Discrete-Time Sinusoidal Signals:**

- Expressed as:

$$x(n) = A \cos(\omega n + \theta) \quad \text{or} \quad x(n) = A \cos(2\pi f n + \theta)$$

where  $n$  is the sample number,  $A$  is the amplitude,  $\omega$  is the frequency in radians per sample, and  $\theta$  is the phase in radians.

- A discrete-time sinusoid is periodic only if its frequency  $f$  is a rational number.
- Discrete-time sinusoids whose frequencies are separated by an integer multiple of  $2\pi$  are identical.
- The highest rate of oscillation is attained when  $\omega = \pi$  or equivalently,  $f = \frac{1}{2}$ .

### Aliasing

- The sinusoids having the frequency  $|\omega| > \pi$  are the alias of the corresponding sinusoids with frequency  $|\omega| < \pi$
- **Aliasing** occurs because periodic sampling of a continuous-time signal maps an infinite range of frequencies into a finite range for the discrete-time signal.
  - Frequencies  $F = F_0 + k F_s$ , where  $k$  is an integer, are indistinguishable from  $F_0$  after sampling and are aliases of  $F_0$ .
  - For example, with a sampling rate  $F_s = 40$  Hz, a 50 Hz signal is an alias of a 10 Hz signal.

### Sampling of Analog Signals

- **Sampling** is the process of selecting values of an analog signal at discrete-time instants.
- The relationship between continuous and discrete time variables is given by:

$$t = nT$$

where  $T$  is the sampling period.

- The reciprocal of  $T$  is the **sampling rate**  $F_s$  or **sampling frequency**:

$$F_s = \frac{1}{T}$$

- The relationship between analog and digital frequencies is given by:

$$f = \frac{F}{F_s}$$

The highest frequency in a discrete-time signal is  $f = \frac{1}{2}$ .

### The Sampling Theorem

- **Shannon's sampling theorem** states that a continuous-time signal with no frequency components higher than  $F_{\max}$  can be completely determined by samples taken at a rate  $F_s \geq 2 F_{\max}$ .
  - The minimum sampling rate  $F_s = 2 F_{\max}$  is called the **Nyquist rate**.
  - To avoid aliasing, the sampling rate must be selected such that  $F_s > 2 F_{\max}$ .

### Quantization of Continuous-Amplitude Signals

- **Quantization** is the process of converting a discrete-time continuous-amplitude signal into a digital signal by expressing each sample as a finite number of digits.
  - The error introduced in this process is called **quantization error** or **quantization noise**, and is given by:

$$e_q(n) = x_q(n) - x(n)$$

- Quantization is an irreversible process because samples within a certain distance of a quantization level are assigned the same value.

### Digital-to-Analog Conversion

- This topic is mentioned but not detailed in the document.
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## CSE\_223\_2.pdf: Introduction to Discrete-Time Signals

### What is a Discrete-Time Signal?

- A **discrete-time signal** is a sequence of values, denoted as  $x(n)$ , where  $n$  represents the time index, typically an integer.
- Unlike continuous-time signals, discrete-time signals are only defined at integer values of  $n$  and are not defined for non-integer values of  $n$ .

### Graphical Representation:

- Discrete-time signals are displayed as a series of **discrete points** on a graph. The time origin ( $n=0$ ) is typically marked with a symbol ( $\uparrow$ ).

### Sampling:

- Discrete-time signals are obtained by sampling an **analog signal** ( $x_a(t)$ ), where  $x(n)$  is defined as  $x_a(nT)$ , with  $T$  being the sampling period.
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## Elementary Discrete-Time Signals

### 1. Unit Sample Sequence (Unit Impulse):

- A signal that is **1** at  $n = 0$  and **0** everywhere else.

### 2. Unit Step Signal:

- A signal that is **0** for  $n < 0$  and **1** for  $n \geq 0$ .

### 3. Unit Ramp Signal:

- A signal whose value increases **linearly** with time for  $n \geq 0$ .

### 4. Exponential Signal:

- Real Exponential:** When the parameter  $a$  is real, the signal is a simple exponential.
  - Complex Exponential:** If  $a$  is complex (i.e.,  $a = re^{j\theta}$ ), the signal is a complex exponential, represented as:  
–  $r^n * e^{j\theta n}$  or equivalently  $r^n(\cos(\theta n) + j\sin(\theta n))$ .
  - The **magnitude** of the complex exponential is  $A(n) = r^n$ , and the **phase** is  $\phi(n) = \theta n$ .
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## Classification of Discrete-Time Signals

### 1. Energy Signals and Power Signals:

- Energy:** The capacity of a signal to create a change.

$$E = \sum_{n=-\infty}^{\infty} |x[n]|^2$$

- Power:** The rate at which energy is used or transmitted.

$$P = \lim_{N \rightarrow \infty} \frac{1}{(2N+1)} \sum_{n=-N}^N |x[n]|^2$$

**Energy Signal:**

- A signal  $\mathbf{x(n)}$  is an **energy signal** if its **total energy  $\mathbf{E}$**  (calculated by summing the squared magnitudes of the signal) is finite ( $\mathbf{0 < E < \infty}$ ).

**Power Signal:**

- The **average power  $\mathbf{P}$**  of a discrete-time signal  $\mathbf{x(n)}$  is calculated as the average of the sum of the squared magnitudes of the signal. If  $\mathbf{P}$  is finite and non-zero, the signal is a **power signal**.

**Key Relationships:**

- If  $\mathbf{E}$  is finite,  $\mathbf{P = 0}$ .
- If  $\mathbf{E}$  is infinite,  $\mathbf{P}$  can be finite or infinite.
- A signal cannot be both an energy signal and a power signal; they are **mutually exclusive**.
- A signal is neither an energy nor a power signal if both energy and power are infinite.
- **Practical Signals:** Most practical signals are energy signals with finite duration and amplitude. Power signals require infinite duration, making them physically impossible to generate.
- Signals with **constant amplitude** over an infinite duration are **power signals**.

**2. Periodic and Aperiodic Signals:****Periodic Signal:**

- A signal  $x(n)$  is periodic with **period  $\mathbf{N}$**  if  $x(n + N) = x(n)$  for all  $\mathbf{n}$ . The smallest such  $\mathbf{N}$  is called the **fundamental period**.

**Aperiodic Signal:**

- A signal is **aperiodic** (or non-periodic) if no such  $\mathbf{N}$  exists.

**3. Symmetric (Even) and Antisymmetric (Odd) Signals:****Symmetric (Even) Signal:**

- A signal  $\mathbf{x(n)}$  is **symmetric** (even) if  $\mathbf{x(-n) = x(n)}$ .

**Antisymmetric (Odd) Signal:**

- A signal  $\mathbf{x(n)}$  is **antisymmetric** (odd) if  $\mathbf{x(-n) = -x(n)}$ .
- Any arbitrary signal can be expressed as the sum of an **even** and an **odd** signal:
  - **Even component:**  $(x(n) + x(-n))/2$

- **Odd component:**  $(x(n) - x(-n))/2$
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## Simple Manipulations of Discrete-Time Signals

### 1. Time Shifting:

- Replacing  $n$  with  $n - k$  shifts the signal in time.
  - If  $k$  is positive, it is a **delay**.
  - If  $k$  is negative, it is an **advance**.

### 2. Time Reversal (Folding):

- Replacing  $n$  with  $-n$  reflects the signal about the time origin ( $n = 0$ ).

### 3. Time Scaling (Down-sampling):

- Replacing  $n$  with  $\mu n$ , where  $\mu$  is an integer, scales the time base.
  - For example,  $y(n) = x(2n)$  represents **down-sampling** the signal.

### 4. Amplitude Modifications:

#### Amplitude Scaling:

- Multiplying the signal by a constant  $A$  scales the amplitude of every sample.

#### Addition of Signals:

- Adding two signals,  $x_1(n)$  and  $x_2(n)$ , results in a new signal  $y(n) = x_1(n) + x_2(n)$ .

#### Multiplication of Signals:

- The product of two signals is calculated sample-by-sample:  $y(n) = x_1(n) * x_2(n)$ .
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## Discrete-Time Systems

### Definition:

- A **discrete-time system** is a device or algorithm that transforms a **discrete-time input signal** into a corresponding **output signal**.

### Input-Output Relationship:

- The system's behavior can be described by the relationship between its **input** and **output** signals.

### Block Diagram Representation:

- Basic building blocks include:
    - **Adder:** Sums two or more signals.
    - **Constant Multiplier:** Multiplies a signal by a constant.
    - **Signal Multiplier:** Multiplies two signals.
    - **Unit Delay Element:** Delays the signal by one sample.
    - **Unit Advance Element:** Advances the signal by one sample.
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## Classification of Discrete-Time Systems

### 1. Static vs. Dynamic Systems:

#### Static (Memoryless) System:

- The output depends only on the current input, not on past or future inputs.

#### Dynamic System:

- The output depends on past or future inputs, implying the system has **memory**.

### 2. Time-Invariant vs. Time-Variant Systems:

#### Time-Invariant System:

- The system's behavior doesn't change with time. Shifting the input by  $k$  samples results in the same shift in the output.

#### Time-Variant System:

- The system's behavior changes with time.

### 3. Linear vs. Nonlinear Systems:

#### Linear System:

- A system that satisfies the **superposition principle** (the output is the sum of the individual responses to each input signal).



**Nonlinear System:**

- A system that does not satisfy the superposition principle.

**4. Causal vs. Noncausal Systems:**

**Causal System:**

- The output depends only on **present** and **past** inputs, not on future inputs.

**Noncausal System:**

- The output depends on **future** inputs.

**5. Stable vs. Unstable Systems:**

**Stable (BIBO) System:**

- A system is stable if **bounded input** results in a **bounded output**.

**Unstable System:**

- If a bounded input leads to an **unbounded** output, the system is unstable.
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## Interconnection of Discrete-Time Systems

- Systems can be interconnected to form larger systems using configurations such as **cascade** (series) and **parallel** arrangements.
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## Analysis of Linear Time-Invariant (LTI) Systems

**Techniques for Analysis:**

- Directly solve the input-output equations or decompose the input signal into elementary components and apply linearity to find the total output.

**Impulse Response:**

- The **impulse response** of an LTI system is the response to an input **( $\delta[n-k]$ )** (the unit sample).

**Convolution Sum:**

- The output  $\mathbf{y}(\mathbf{n})$  of an LTI system is given by the **convolution sum** of the input signal  $\mathbf{x}(\mathbf{n})$  and the system's impulse response  $\mathbf{h}(\mathbf{n})$ :
    - $y(n) = \sum_{k=-\infty}^{\infty} x(k)h(n-k)$
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## Correlation of Discrete-Time Signals

**Objective:**

- To measure the degree of similarity between two signals.

**Applications:**

- Used in areas such as radar, sonar, digital communications, and geology.

**Crosscorrelation and Autocorrelation:**

- Methods for measuring similarity between signals.
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