

Digital Signal Processing

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> http://www.engineering.uodiyala.edu.iq/ https://www.facebook.com/Engineer.College1



Course Overview

- Signals and Systems.
- Classification of Discrete Digital Systems.
- Time and Frequency Domains Analysis.
- Signal Transformation Methods: Fourier, Wavelet and Z-transform.
- Digital Filter Types: FIR and IIR Filters.
- Digital Filter Design.
- Analog Filter Design.
- DSP Applications.



Books

- J.G. Proakis and D.G. Manolakis, **Digital Signal Processing**, <u>4rd edition</u>, Prentice-Hall, 2006.
- R.G Lyons, **Understanding Digital Signal processing**, <u>3rd edition</u>, Prentice-Hall, (Amazon's top-selling for five straight year) ,2011.
- Monsons Hays, Schaums Outline of Digital Signal processing, <u>2nd edition</u> ,McGraw-Hill Companies, 2012.
- Richard, **The Essential Guide to Digital Signal Processing**, <u>1st edition</u> Prentice-Hall ,ePUB, 2014.
- J.G. Proakis , **Digital Signal Processing Using MATLAB**, <u>3rd edition</u>, Cengage Learning , 2012.



- **Signal**: It can be broadly defined as any physical quantity that varies as a time and/or space and has the ability to convey information, examples of these signals are:
 - Electrical signals: currents and voltages in AC circuits, radio communications signals, video signals etc.
 - Mechanical signals: sound or pressure waves, vibrations in a structure, earthquakes, etc.
 - **Biomedical signals**: electro-encephalogram, lung and heart monitoring, X-ray and other types of images.
 - Finance: time variations of a stock value or a market index.

Digital Signal: operating by the use of discrete signals to represent data in the form of numbers.



Processing: a series operations performed according to programmed instructions.



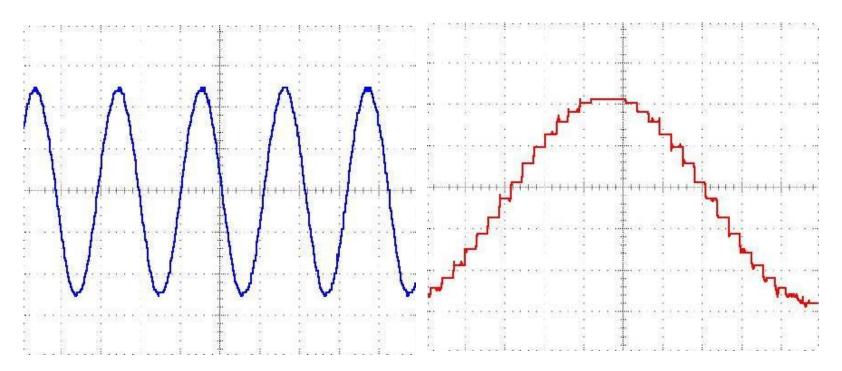
changing or analysing information which is measured as discrete sequences of numbers

"Learning digital signal processing is not something you accomplish; it's a journey you take".

R.G Lyons, Understanding Digital Signal processing

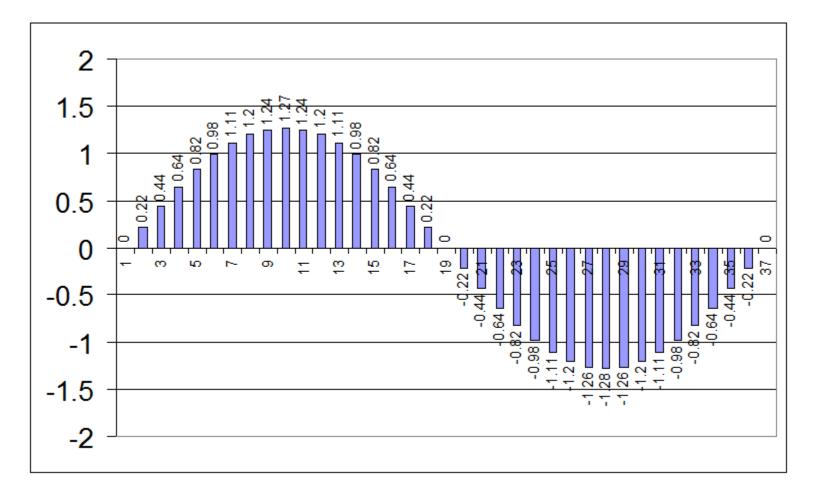


Converting a continuously changing waveform (analog) into a series of discrete levels (digital)



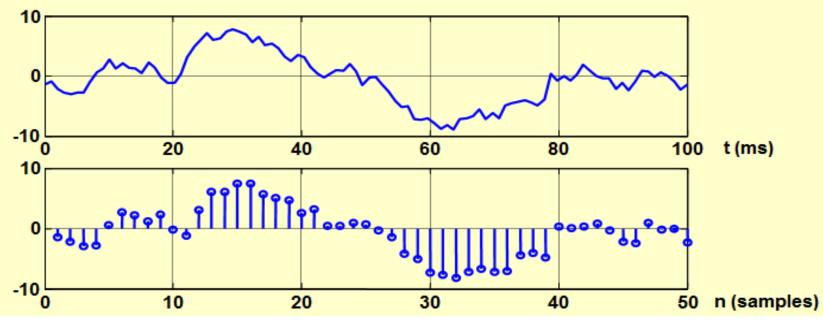
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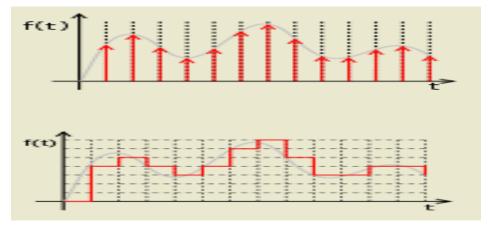




- Discrete-time signals are represented by sequence of numbers
 - The nth number in the sequence is represented with x[n]
- Often times sequences are obtained by sampling of continuous-time signals
 - In this case x[n] is value of the analog signal at $x_c(nT)$
 - Where T is the sampling period







discrete signal is discrete in time but continuous in amplitude.

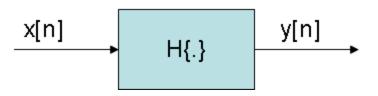
digital signal is discrete in both time and amplitude.



- The most convenient mathematical representation of a signal is via the concept of a function, say x(t). In this notation:
 - **x** represents the dependent variable (e.g. voltage, pressure, etc.)
 - *t* represents the independent variable (e.g. time, space, etc.).
- Depending on the nature of the independent and dependent variables, different types of signals can be identified such as:
 - ✓ Analog signals
 - ✓ Discrete signals
 - ✓ Digital signals
 - ✓ Multi-channel signals
 - ✓ Multi-dimensional signals



- **System**: A physical entity that operates on a set of primary signals (the inputs) to produce a corresponding set of resultant signals (the output).
- The operations, or processing, may take several forms: modification, combination, decomposition, filtering, extraction of parameters, etc.
- System characterization: a system can be represented mathematically as a transformation between two signal sets , as in x[n] → y[n]:





- Depending on the nature of the signals on which the system operates, different basic types of systems may be identified:
 - Analog or continuous-time system: the input and output signals are analog in nature.
 - Discrete-time system: the input and output signals are discrete.
 - Digital system: the input and outputs are digital.
 - Mixed system: a system in which different types of signals (i.e.analog,discrete and/or digital) coexist.



Discussion:

- Early education in engineering focuses on the use of calculus to analyze various systems and processes at the analog level:
 - motivated by the prevalence of the analog signal model
 - e.g.: circuit analysis using differential equations
- Yet, due to extraordinary advances made in micro-electronics, the most common/powerful processing devices today are digital in nature.
- Thus, there is a strong, practical motivation to carry out the processing of analog real-world signals using such digital devices.
- This has lead to the development of an engineering discipline know as digital signal processing DSP.

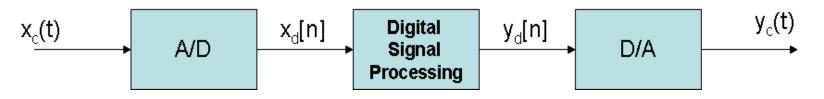
Digital Signal Processing:

In its most general form, DSP refers to the processing of analog signals by means of discrete-time operations implemented on digital hardware.



Basic components of a DSP System

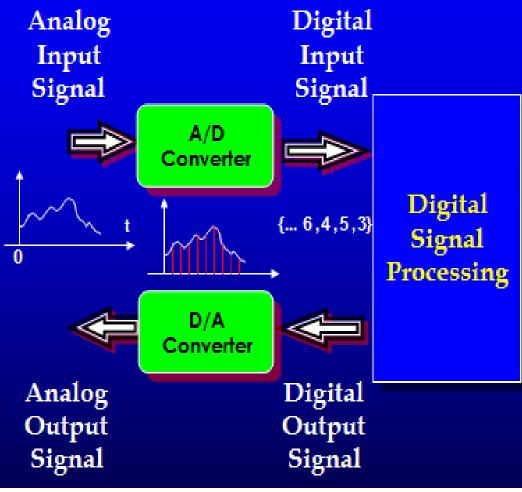
• In its most general form, a DSP system will consist of three main components, as illustrated below:



- The analog-to-digital (A/D) converter transforms the analog signal $x_c(t)$ at the system input into a digital signal $x_d[n]$.
- The digital system performs the desired operations on the digital signal x_d [n] and produces a corresponding output y_d [n] also in digital form.
- The digital-to-analog (D/A) converter transforms the digital output $y_d[n]$ into an analog signal $y_c(t)$ suitable for interfacing with the outside world.



Basic components of a DSP System



A/D Converter

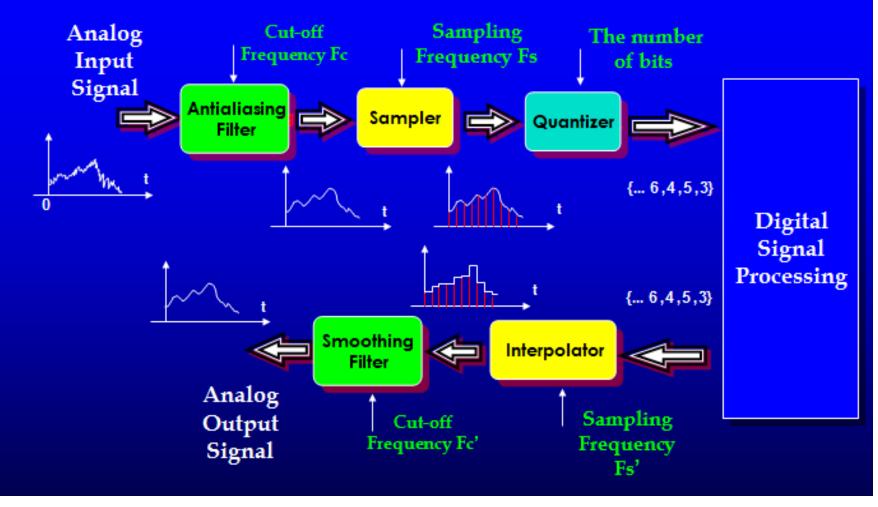
Converts an analog signal into a sequence of digits

D/A Converter⁽

Converts a sequence of digits into an analog signal



Basic components of a DSP System

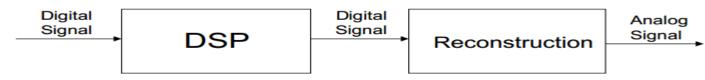




To implement DSP we must be able to:



(1) perform numerical operations including, for example, additions, multiplications, data transfers and logical operations either using computer or special-purpose hardware.



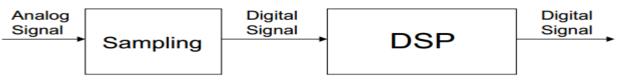
(2) convert the digital information, after being processed back to an analog signal – involves digital-to- analog conversion and reconstruction .

e.g. text-to-speech signal (characters are used to generate artificial sound)



DSP Implementation –Analog/Digital Conversion

To implement DSP we must be able to:



3) convert analog signals into the digital information - sampling & involves analog-todigital conversion.

e.g. Touch-Tone system of telephone dialling (when button is pushed two sinusoid signals are generated (tones) and transmitted, a digital system determines the frequencies and uniquely identifies the button – digital (1 to 12) output



perform both A/D and D/A conversions

e.g. digital recording and playback of music (signal is sensed by microphones, amplified, converted to digital, processed, and converted back to analog to be played.



DSP Chips : Special Purpose Hardware

Introduction of the microprocessor in the late 1970's and early 1980's meant DSP techniques could be used in a much wider range of applications.





Bluetooth headset



Household appliances



Home theatre system

DSP chip – a *programmable* device, with its own native instruction code

designed specifically to meet numerically-intensive requirements of DSP

capable of carrying out millions of floating point operations per second

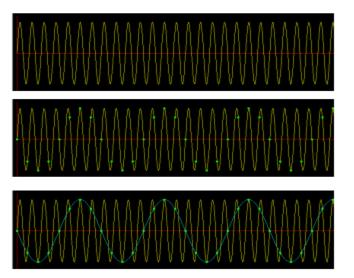


Limitations of DSP-Aliasing

Most signals are analog in nature, and have to be sampled.

loss of information

we only take samples of the signals at intervals and don't know what happens in between



Gjendemsjø, A. Aliasing Applet, Connexions, http://cnx.org/content/m11448/1.14



cannot distinguish between higher and lower frequencies

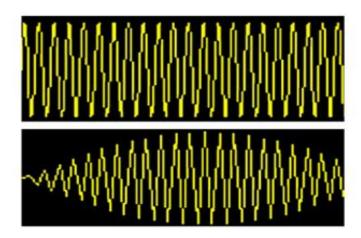
(recall from 1B Signal and Data Analysis)

Sampling theorem: to avoid aliasing, sampling rate must be at least twice the maximum frequency component (`bandwidth') of the signal



Limitations of DSP - Antialiasing Filter

 Sampling theorem says there is enough information to reconstruct the signal, which does not mean sampled signal looks like original one



Each sample is taken at a slightly earlier part of a cycle correct reconstruction is not just connecting samples with straight lines

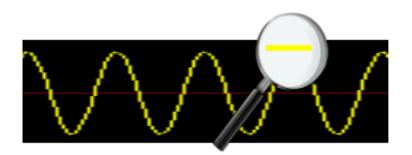
needs antialias filter (to filter out all high frequency components before sampling) and the same for reconstruction – it does remove information though



Limitations of DSP – Frequency Resolution

Most signals are analog in nature, and have to be sampled loss of information

we only take samples for a limited period of time



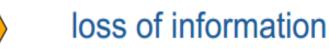
limited frequency resolution

does not pick up "relatively" slow changes

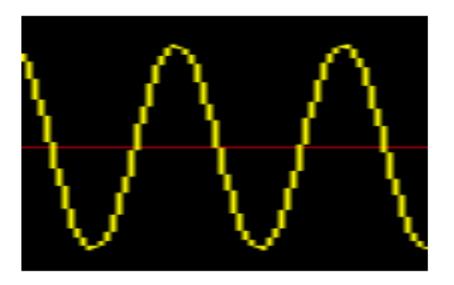


Limitations of DSP – Quantization Error

Most signals are analog in nature, and have to be sampled



 limited (by the number of bits available) precision in data storage and arithmetic





smoothly varying signal represented by "stepped" waveform



Advantages of Digital over Analog Signal Processing

Why still do it?

- Digital system can be simply reprogrammed for other applications / ported to different hardware / duplicated (Reconfiguring analog system means hardware redesign, testing, verification)
- DSP provides better control of accuracy requirements (Analog system depends on strict components tolerance, response may drift with temperature)
- Digital signals can be easily stored without deterioration (Analog signals are not easily transportable and often can't be processed offline)
- More sophisticated signal processing algorithms can be implemented (Difficult to perform precise mathematical operations in analog form)



Advantages of Digital over Analog Signal Processing

Advantages:

- Robustness:
- Signal levels can be regenerated.
- Precision not affected by external factors
- Storage capability:
- DSP system can be interfaced to low-cost devices for lasting storage
- allows for off-line computations
- Flexibility:
- Easy control of system accuracy via changes in sampling rate and number of representation bits.
- Software programmable \rightarrow reconfiguring the DSP operations simply by changing the program.
- Structure:
- Easy interconnection of DSP blocks (no loading problem)
- Possibility of sharing a processor between several tasks

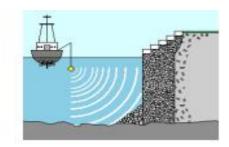
Disadvantages:

- Cost/complexity added by A/D and D/A conversion.
- Input signal bandwidth is technology limited.
- Quantization effects.



Applications of DSP-Radar

Radar and Sonar:



Examples

1) target detection – position and velocity estimation

2) tracking



Applications of DSP: Biomedical

Biomedical: analysis of biomedical signals, diagnosis, patient monitoring, preventive health care, artificial organs



Examples:

 electrocardiogram (ECG) signal – provides doctor with information about the condition of the patient's heart

2) electroencephalogram (EEG) signal – provides Information about the activity of the brain





Applications of DSP: Speech

Speech applications:

Examples



1) noise reduction – reducing background noise in the sequence produced by a sensing device (microphone)



2) speech recognition – differentiating between various speech sounds

 synthesis of artificial speech – text to speech systems for blind





Applications of DSP: Communications

Communications:



Examples

telephony – transmission of information in digital form via



telephone lines, modern technology, mobile phones

2) encoding and decoding of the information sent over a physical channel (to optimise transmission or to detect or correct errors in transmission)





Applications of DSP: Image Processing

Image Processing:

Examples

 content based image retrieval – browsing, searching and retrieving images from database



INFORMATION RETRIEVAL



2) image enhancement

2) compression - reducing the redundancy in the image data to optimise transmission / storage





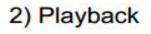
Applications of DSP: Music

Music Applications:



Examples:

1) Recording







3) Manipulation (mixing, special effects)



Applications of DSP: Multimedia

Multimedia:



generation storage and transmission of sound, still images, motion pictures

Examples:

1) digital TV





2) video conferencing



Applications of DSP



