

Digital Signal Processing (Lecture Note 1)

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EE 401

Course Descriptions

- Techniques of modern digital signal processing
- Theory of Fourier series, Fourier transform, signal sampling, Z-transform
- Design and implementation of digital filters
- Application of digital signal processing techniques

Tentative Syllabus

Week 1: Overview of typical measurement systems and their nature in aspects of signal processing

Week 2: Characteristics of linear time invariant systems

Week 3: Fourier series

Week 4: Fourier transform

Week 5: Sampling theory and Discrete time Fourier transform

Week 6: Discrete Fourier transform and data windowing

Week 7: Data windowing

Week 8: Midterm Exam

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Tentative Syllabus

Week 9: Z transform

Week 10: Basic concepts of filtering

Week 11: Design of finite impulse response (FIR) filters

Week 12: Design of infinite impulse response (IIR) filters

Week 13: Structures for discrete-time systems

Week 14: Short-time Fourier transform

Week 15: Spectral analysis

Week 16: Final Exam

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Grading Policy and Text Book

- Midterm Exam: 35%
- Final Exam: 35%
- Homework: 30%
- **Discrete-time Signal Processing**

(Third Edition, Alan V. Oppenheim & Ronald W. Schafer)

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Typical Measurement Systems

What is a signal?

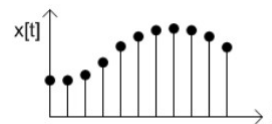
- Speech
- Biosignals
- A photographic image

└─ Continuous, Discrete

1. Continuous-time signal



2. Discrete-time signals

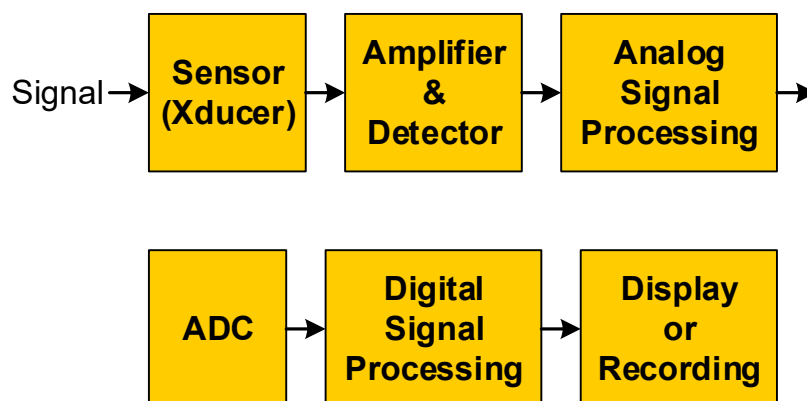


Sequences of numbers

$X = \{x[n]\}$, n is an integer

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Typical Measurement Systems



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Sensor (Transducer)

- A transducer is a device that converts energy from one form to another.
 - Input: non-electrical energy
 - Output: voltage (or current) whose amplitude is proportional to the measured energy.
- A transducer frequently establishes the major performance criterion of a measurement system.
- A good system design usually calls for care in the choice or design of transducer elements.
 - An efficient, low-noise transducer design can often reduce the need for extensive subsequent signal processing and still produce a better measurement.

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Analog Signal Processing

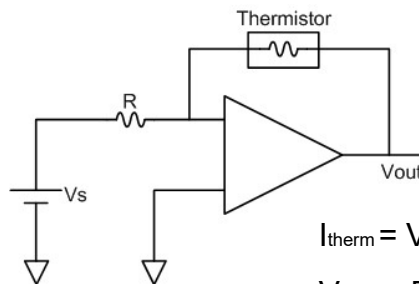
- Analog signal processing functions are usually placed in between a transducer and ADC (analog-to-digital converter).
- The first analog stage depends on the basic transducer operation.
 - If the transducer is based on a variation in electrical property, the first stage must convert that variation in electrical property into a variation in voltage.
 - If the transducer element is a voltage generator, the first stage is usually an amplifier.
 - If the transducer produces a current output, as is the case in many electromagnetic detectors, then a current-to-voltage amplifier (transconductance amplifier) is used to produce a voltage output.

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Example

- A device with an electrical resistance that decreases rapidly as its temperature rises, used in thermometers.

$$V_{out} = I(Z + \Delta Z) \text{ where } \Delta Z = f(\text{input energy})$$



$$I_{therm} = V_s/R = \text{constant}$$

$$V_{out} = [(R_{therm} + \Delta R_{therm})/R] \cdot V_s$$

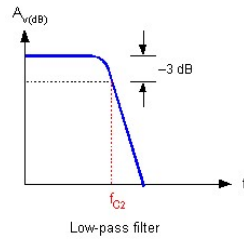
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Analog Filters

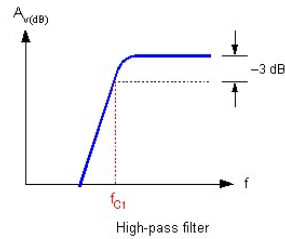
- The analog signal processing circuitry will usually contain some filtering, both to remove noise and appropriately condition the signal for ADC.
 - All transducers and electronics produce some noise such as thermal noise from the transducer and other electronic components, flicker (or burst, or $1/f$) noises from electronic components.
 - Thermal noise acting as a white noise usually contains a wide range of frequencies, and the burst noise has large spectral components in the dc and low frequency range.
 - Band-pass filtering removes those noise.
 - Low-pass or band-pass filtering makes the bandwidth of signals limited, thus making sampling error minimized in performing analog-to-digital conversion.

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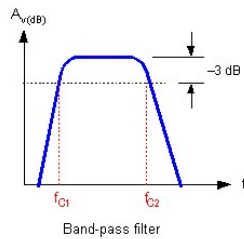
Filter Types



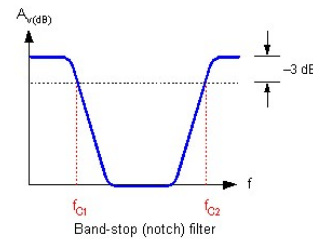
Low-pass filter



High-pass filter



Band-pass filter

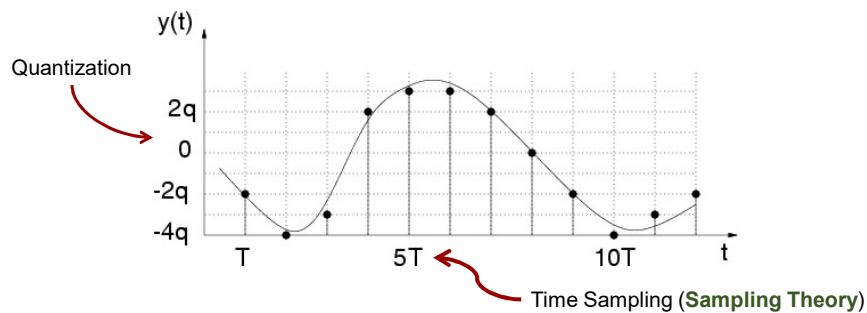


Band-stop (notch) filter

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Analog-to-Digital Conversion

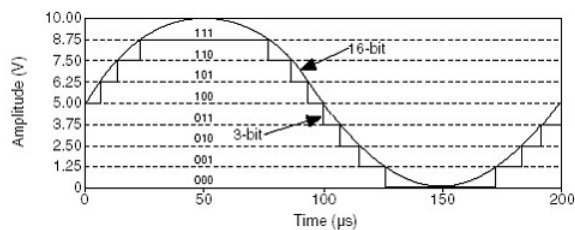
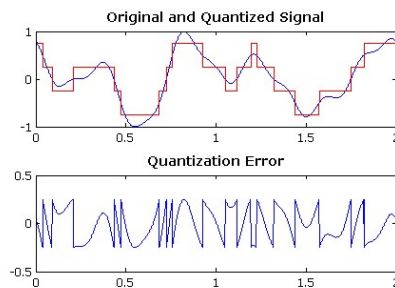
- ADC converts an analog voltage to an equivalent digital number.
 - To convert a continuous waveform to digital format requires slicing the signal in two ways: slicing in time and slicing in amplitude.



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Quantization Error

- The number of bits used for conversion sets a lower limit on the resolution, and also determines the quantization error that can be thought of as a noise process added to the signal.
- 8-Bit vs. 12-Bit ADC
 - 12-Bit ADC has less quantization error than 8-Bit ADC



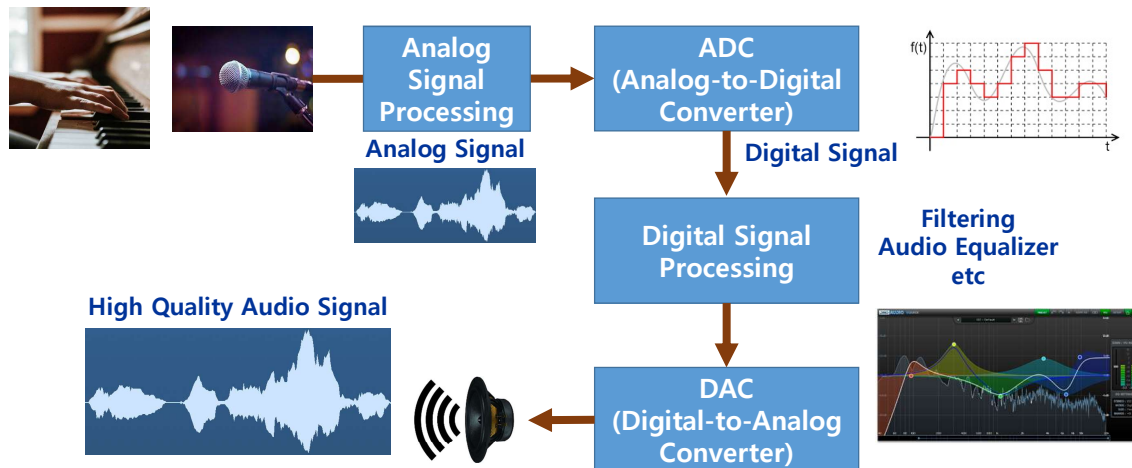
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Digital Signal Processing

- Digital Filtering
 - Based on discrete Fourier transform (DFT), spectral analysis, z-transform, and Convolution
 - Signal separation and noise reduction
- Signal Enhancement
- Feature Extraction
- etc

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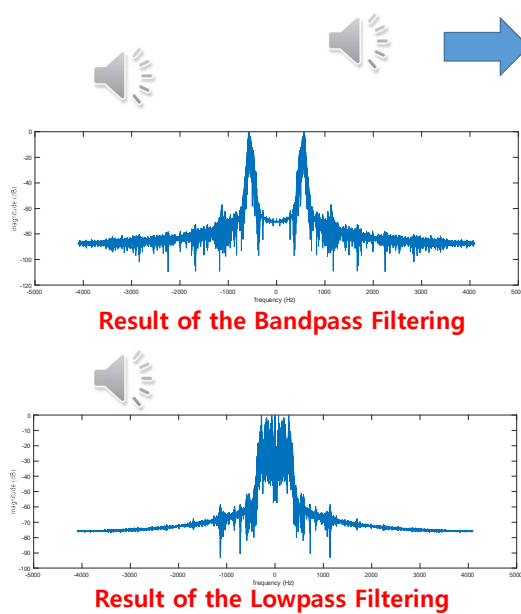
Example: Audio Signal Processing



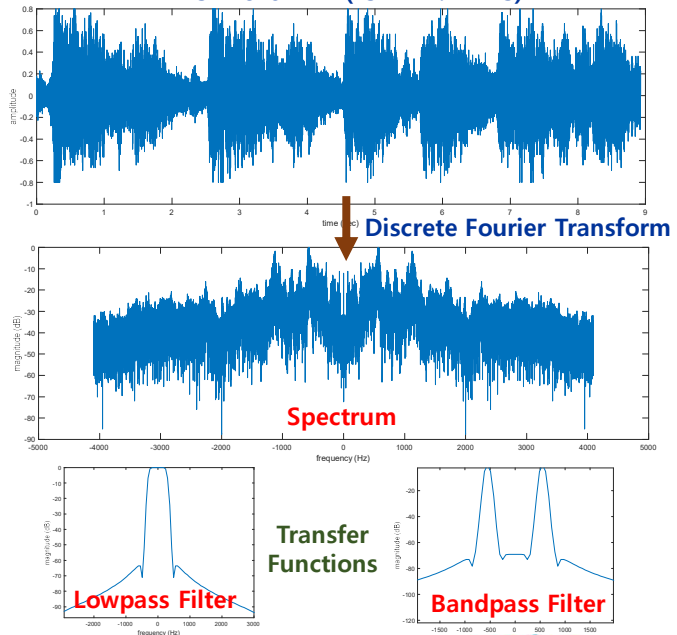
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Digital Signal - Handel



$F_s = 8192 \text{ Hz}$ ($T_s \sim 12.21 \text{ ms}$)



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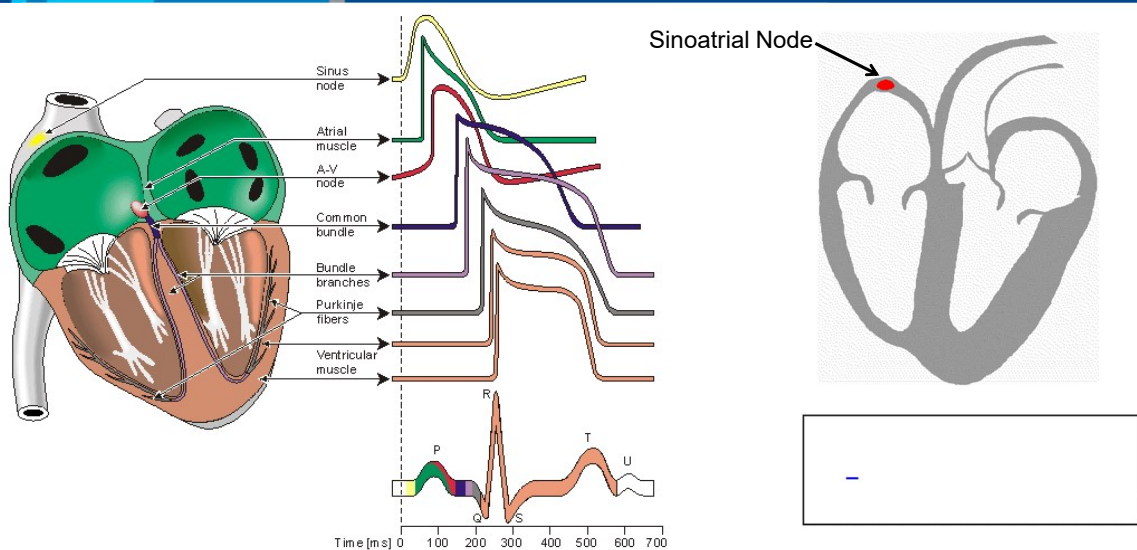
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Example: ECG as a Biosignal

- A method to record bioelectric currents generated by the heart.
 - A routine investigation in patients who complain of symptoms such as chest pain and breathlessness.
 - Diagnosis of abnormal heart rhythms, i.e., heart disease: ischemic heart disease, hypertrophy patterns, atrioventricular (AV) block, etc.
- Electrical impulses in the heart begin from the sinoatrial node and travel through the intrinsic conducting system to the heart muscle.
 - The impulses stimulate the myocardial muscle fibers to contract, which is called "systole".

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A Normal ECG Signal



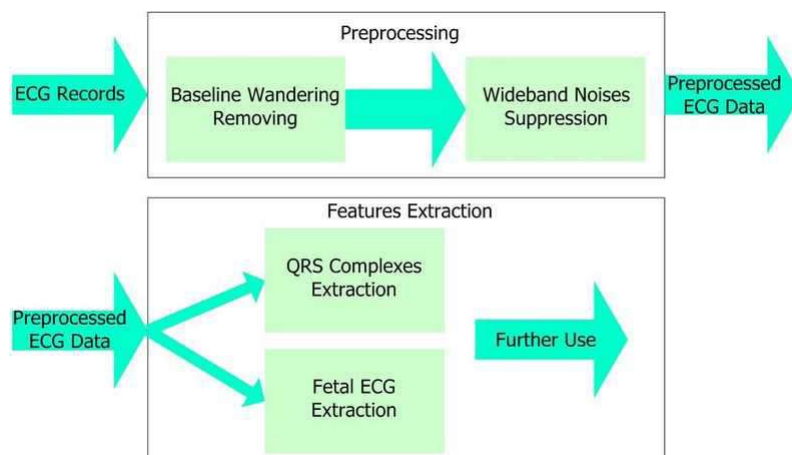
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Signal Processing for ECG

- Several electrodes (or leads) are used to acquire the bioelectric currents.
 - Generally noise and artifacts contaminate ECG signal because those unwanted signals are within the frequency band of interest.
- In order to extract useful information from the noisy ECG signals, signal processing should be conducted.
 - Preprocessing to remove or suppress noise from the raw ECG signals
 - ✓ Baseline wandering removing and wideband noises suppression
 - Feature extraction processing to obtain diagnostic information from the preprocessed ECG signals
 - ✓ QRS complexes detection, fetal heart rate extraction, and etc.

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Typical ECG Signal Processing Flowchart

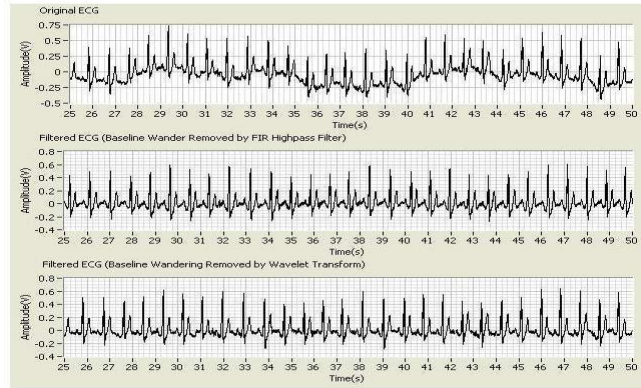


From <http://zone.ni.com/devzone/cda/tut/p/id/6349>

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Preprocessing (1)

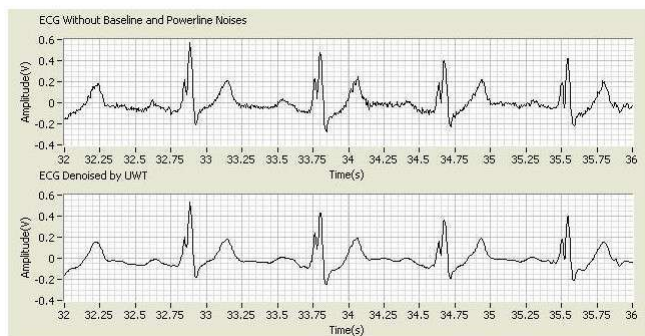
- Baseline wandering, a drift of a running average of the received signal power, can be suppressed by using
 - A high-pass digital filter because the signal consists of very low frequency components between 0.15 and 0.3 Hz.
 - The wavelet transform to eliminate the trend of the ECG signal



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Preprocessing (2)

- Removing wideband noise can be done by using
 - Conventional digital filters
 - Wavelet transform method in which the ECG signal is first decomposed into several subbands, and then each wavelet coefficient is modified by applying a threshold or shrinkage function, and finally the denoised signal is reconstructed.



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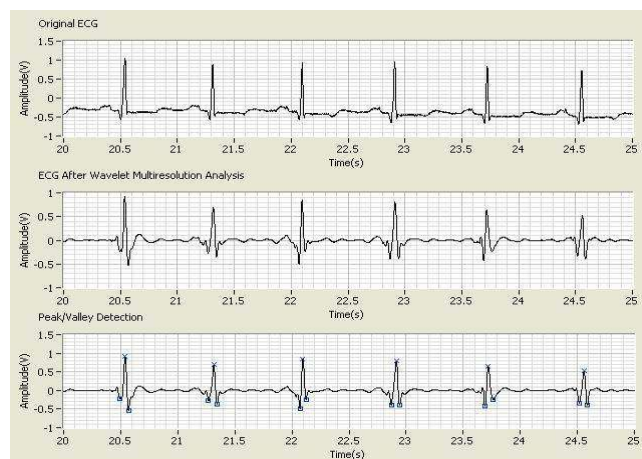
Feature Extraction: QRS complex detection as an Example

- For the purpose of diagnosis, you frequently need to extract various features from the preprocessed ECG data including QRS intervals, QRS amplitudes, PR intervals, ST intervals, and etc.
- The detection of the QRS complexes in an ECG signal is carried out by the detection of the R-peaks.
 - This provides information about the heart rate, the conduction velocity, the condition of tissues within the heart as well as various abnormalities.
 - The presence of noise and time-varying morphology makes the detection difficult.
 - Wavelet-based detection methods can be used to extract the feature from noisy signals by multi-resolution analysis.

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QRS Complex Detection

- This figure shows the ECG signals processed by wavelet multi-resolution analysis and peak/valley detection.
 - Wavelet multi-resolution analysis made the peaks and valleys more distinct.



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Conclusions

ADC (Analog-to-Digital Converter)

- Sampling Theory
 - ✓ Selection of a sampling interval to avoid distortion
- Digital Image Processing
- Audio Signal Processing
- Biomedical Signal Processing
- Basic Knowledge for Deep Learning
- etc

Basic Digital Signal Processing

- Filter Design
 - ✓ Based on Fourier transform
 - ✓ Based on spectrum analysis
- Filter Implementation
 - ✓ Based on z-transform

Digital Signal Processing
Depending on Applications