

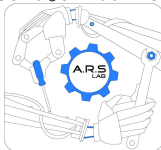
# Basics of Digital Signal Processing

Corrado Santoro

**ARSLAB - Autonomous and Robotic Systems Laboratory**

Dipartimento di Matematica e Informatica - Università di Catania, Italy

santoro@dmi.unict.it



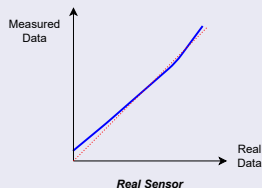
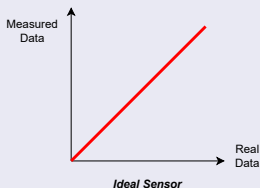
Robotic Systems

# Measures and Errors

- Measure of state variables is a fundamental aspect of control systems
- If the measure is not precise or affected by a significant amount of noise, the whole control system cannot work properly
- However **any measurement** system is always affected by **errors**
- **Measurement errors** have different characteristics and depends on the kind of sensor used

## Sensor Characteristics

- The behaviour of a sensor is in general represented by its **characteristic curve**
- It is plot over a  $XY$  chart that reports in the  $X$  axis the **real data** and in the  $Y$  axis the **measured data**
- For an **ideal** sensor, the characteristic is a 45-degrees straight line
- But for a **real** sensor, the characteristic is a **curve** the is **close** to the 45-degrees straight line



# Measures and Errors

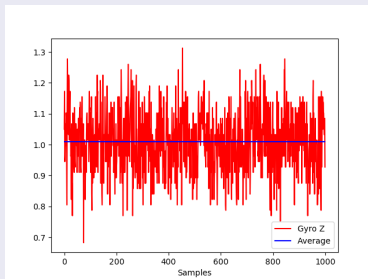
## Sensor Errors

- **Offset:** it is the **non-zero value** given by the sensor when the real data is zero
- **Non-linearity:** it is the difference between the ideal and real characteristic
- **Noise:** it the variation of the measured data when the real data is **constant**
- **Offset** and **Non-linearity** can be reduced by means of **sensor calibration**
- **Noise** (that is harder to be removed) can be reduced by means of **digital filters**



## Noise Errors

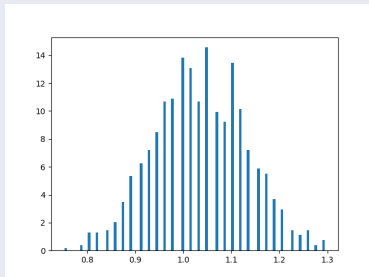
- Here is a plot of the data sampled by a gyroscope, Z-axis, when the system is **stopped**



- The **real data** should be **zero**, but we have an **offset** and a certain amount of **noise**
- The **blue line** is the **average**, if we subtract it, we can remove the offset but not the noise

## Noise Characteristics

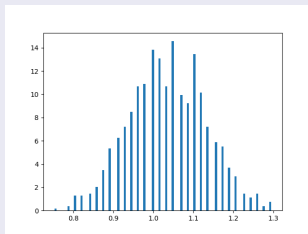
- If we plot the **histogram** of sampled data, we obtain the following chart



- Here we have, on **X axis**, the **values of a sampled data**, and, on **Y axis**, the **number of times** that value is got (data is organized in subintervals)
- The plot is the classical **Gaussian Curve** or **Normal Distribution** that is a common characteristic of noise errors

## Noise Characteristics

- The plot is the classical **Gaussian Curve** or **Normal Distribution** that is a common characteristic of noise errors

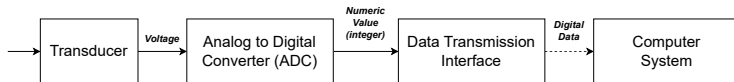


- This curve is characterised by the **average** and the **variance**
- Given that we have  $N$  measures, and given  $x_i$  the measures, we have:

$$\bar{x} = \frac{1}{N} \sum_i x_i$$

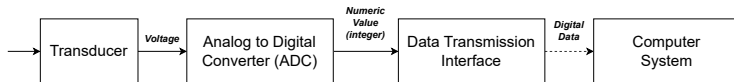
$$\sigma_x^2 = \frac{1}{N} \sum_i (x_i - \bar{x})^2$$

## The Measurement Chain



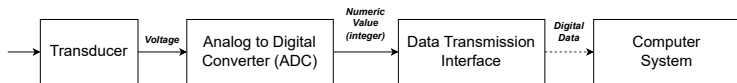
- **Transducer:** a electrical or mechanical system that converts the measure into an electrical magnitude (generally voltage)
- **ADC (Analog-to-Digital Converter):** an electronic circuit that converts a voltage signal into an  $n$ -bit binary representation
- **Data Interface:** a digital circuit (sometimes also CPU-equipped) that performs ADC data conversion and trasmission to a computer system

## The Measurement Chain



- **Range:** the min and max value of the measure that the transducer is able to sample
- **Resolution:** the number of bits of the ADC, or the minimum variation of the measure that the ADC is able to detect
- **Sampling time:** the time required to perform a complete measure, include transducer measuring time and ADC conversion time

## Resolution and Quantization Error

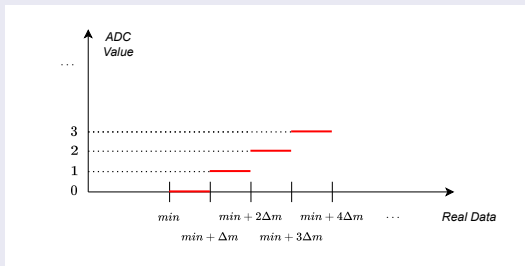


- Due to ADC, the measure value is **discretised** and the minimum variation that can be perceived is:

$$\Delta m = \frac{\text{max} - \text{min}}{2^n - 1}$$

- Where **min** and **max** represent measure range, and **n** the number of bits of the ADC
- Therefore, values allowed are  $\{\text{min}, \text{min} + \Delta m, \text{min} + 2\Delta m, \text{min} + 3\Delta m, \text{min} + 4\Delta m, \dots\}$

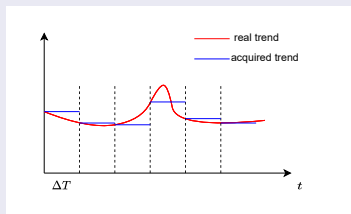
## Resolution and Quantization Error



- By plotting the real value of the measure and the ADC number obtained the result is a **stairway** not a *straight line* (as we expect due to the proportional relationship)
- The phenomenon that produces that plot is called **quantization error**, and strongly depends on the resolution of the ADC

# Sampling Time

## Choice of Sampling Time



- Another important factor of errors is the choice of **sampling time/interval  $\Delta T$**
- Given that any sensor has its **measuring time**, the choice of the *sampling interval* is a design choice
- The choice is made on the basis of what kind of **signal variations** we want to catch
- If  **$\Delta T$  is too low**, the risk is to ignore important variations
- If  **$\Delta T$  is too high**, the risk is to catch too noise
- What is the **right choice**?

## The Fourier Series

Given any (periodic) signal of period  $T$ , it can be expressed as a linear combination of infinite **sin** and **cos** terms:

$$f(t) = \frac{a_0}{2} + \sum_{n=1}^{\infty} [a_n \cos(\frac{2\pi}{T} nt) + b_n \sin(\frac{2\pi}{T} nt)]$$

- $a_n, b_n$  are coefficients that represent the **contribution**, to the overall signal, of the **sinusoid** at frequency  $\frac{n}{T}$
- They constitute the **spectrum** of the signal, i.e. a representation of the magnitude of the **frequency components**

## Spectrum and Noise

- In any signal, the **noise** is a **high frequency** component in a spectrum, so it can be **filtered**
- But if the **signal varies fast**, it also provides high frequency components
- So it is important to understand when **valid data terminate** and **noise begins**
- If valid data end at a frequency  $\bar{f}$ , then set your sampling time at:

$$\Delta T = \frac{1}{2\bar{f}}$$

## Simple Average

- Given  $x$  your sensor variable...
- For each  $\Delta T$ , acquire  $N$  samples  $x'_i$  and compute the mean:

$$x = \frac{1}{N} \sum_{i=1}^N x'_i$$

- If  $t_s$  is the sensor's sampling time, it must be  $N t_s < \Delta T$

## Mobile Average

- Given  $x$  your sensor variable...
- For each  $\Delta T$ , acquire **one** sample  $x'_i$  (at instant  $i$ ), then compute the average of the last  **$N$  past samples**:

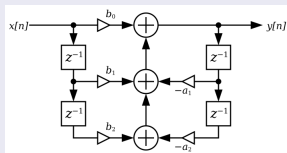
$$x = \frac{1}{N} \sum_{k=0}^{N-1} x'_{i-k}$$

## Digital Filters

- A **digital filter** is discrete dynamic system characterised by the following relation:

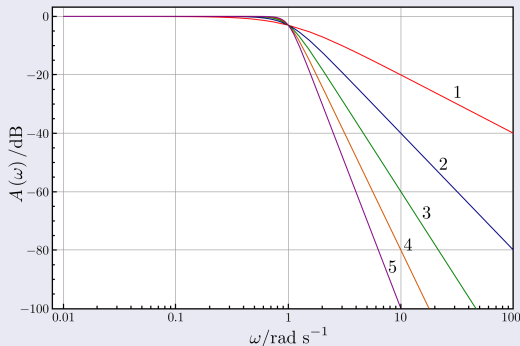
$$y(k) = -a_1y(k-1) - a_2y(k-2) + \dots - a_my(k-m) + b_0u(k) + b_1u(k-1) + \dots + b_nu(k-n)$$

- $u(k)$ , input;  $y(k)$ , output
- $\max(n, m)$  **filter order**
- Coefficients  $a_i, b_i \in [0, 1] \subset \mathcal{R}$  determine the **filter type** and the **noise cut capability**
- The filter behaves as a **weighted average** of past inputs and outputs



## Low-pass Filters and Orders

- A **low pass filter** is a filter that removes data at a frequency **higher than** the **cut-off freq**
- The **order** determines the **slope** of the cut line: the higher the more selective



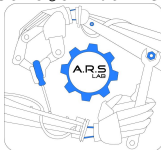
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