

Embedded System Hardware

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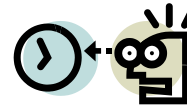
Motivation

(see lecture 1): "The development of ES cannot ignore the underlying HW characteristics. Timing, memory usage, power consumption, and physical failures are important."

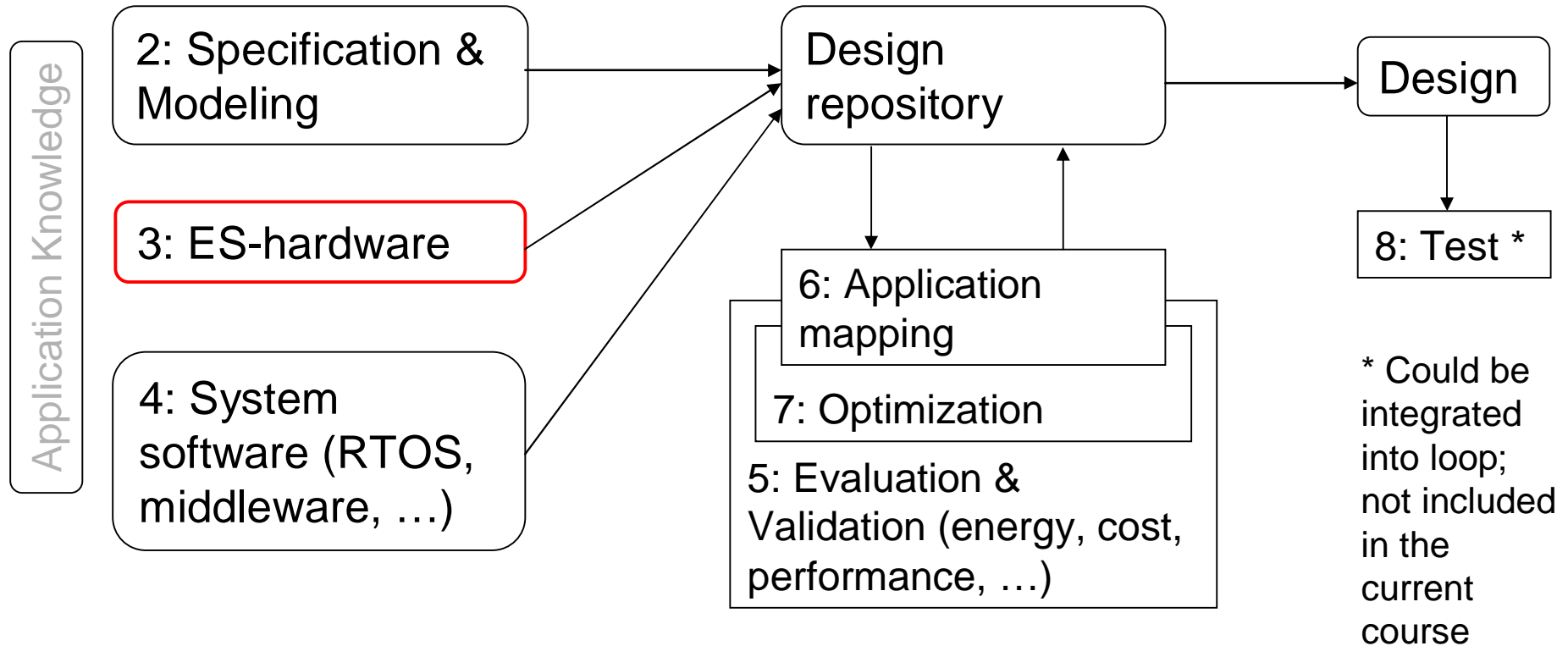
$$\int P dt$$

Reasons for considering hard- and software:

- Real-time behavior
- Efficiency
 - Energy
 - ...
- Security
- Reliability
- ...



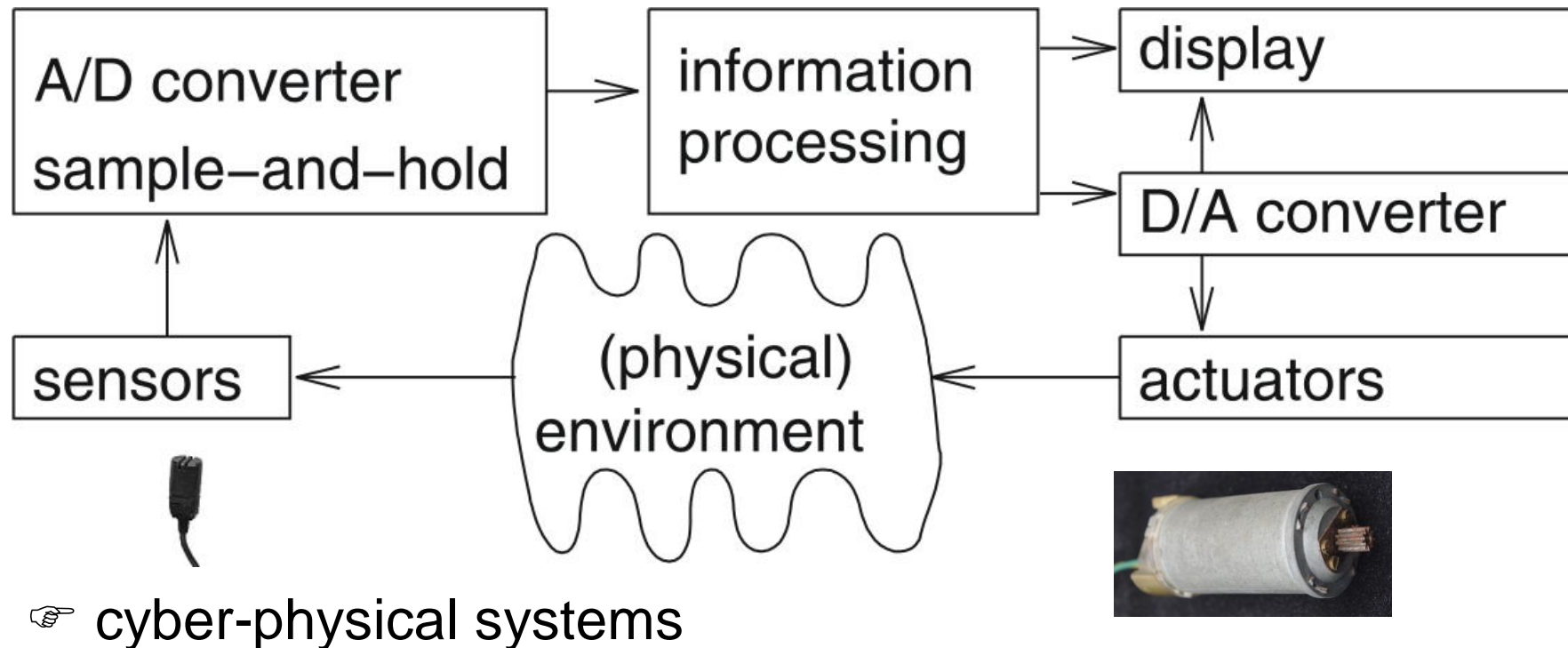
Structure of this course



Generic loop: tool chains differ in the number and type of iterations
Numbers denote sequence of chapters

Embedded System Hardware

Embedded system hardware is frequently used in a loop ("**hardware in a loop**"):



👉 cyber-physical systems

Many examples of such loops

- Heating
- Lights
- Engine control
- Power supply
- ...
- Robots



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Sensors

Processing of physical data starts with capturing this data. Sensors can be designed for virtually every physical and chemical quantity, including

- weight, velocity, acceleration, electrical current, voltage, temperatures, and
- chemical compounds.

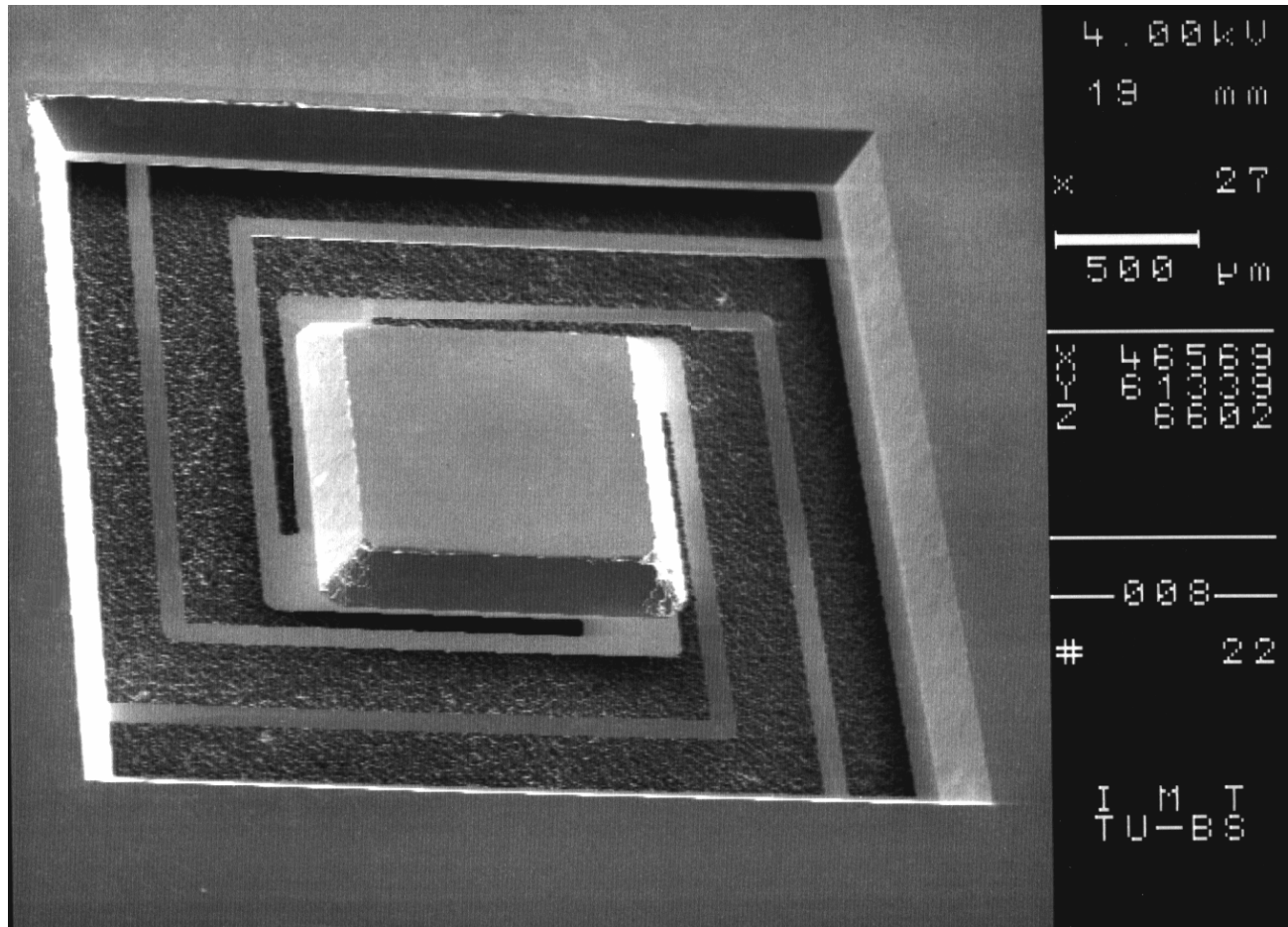
Many physical effects used for constructing sensors.

Examples:

- law of induction (generat. of voltages in a magnetic field),
- light-electric effects.

Huge amount of sensors designed in recent years.

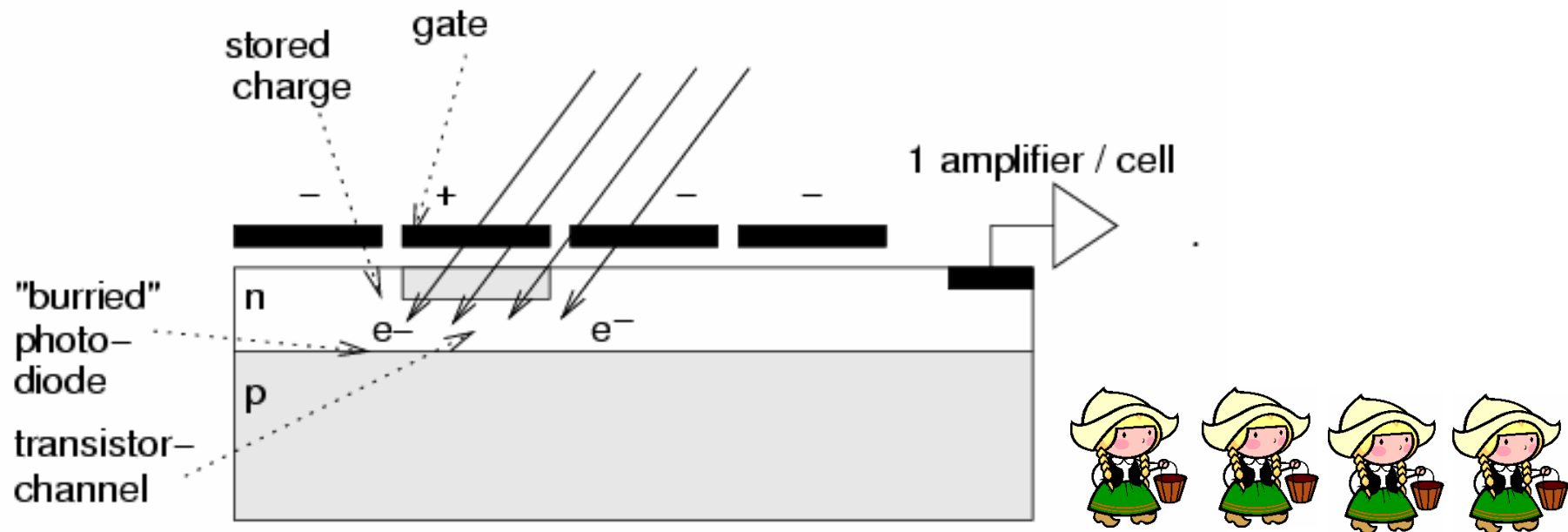
Example: Acceleration Sensor



Courtesy & ©: S. Bütgenbach, TU Braunschweig

Charge-coupled devices (CCD) image sensors

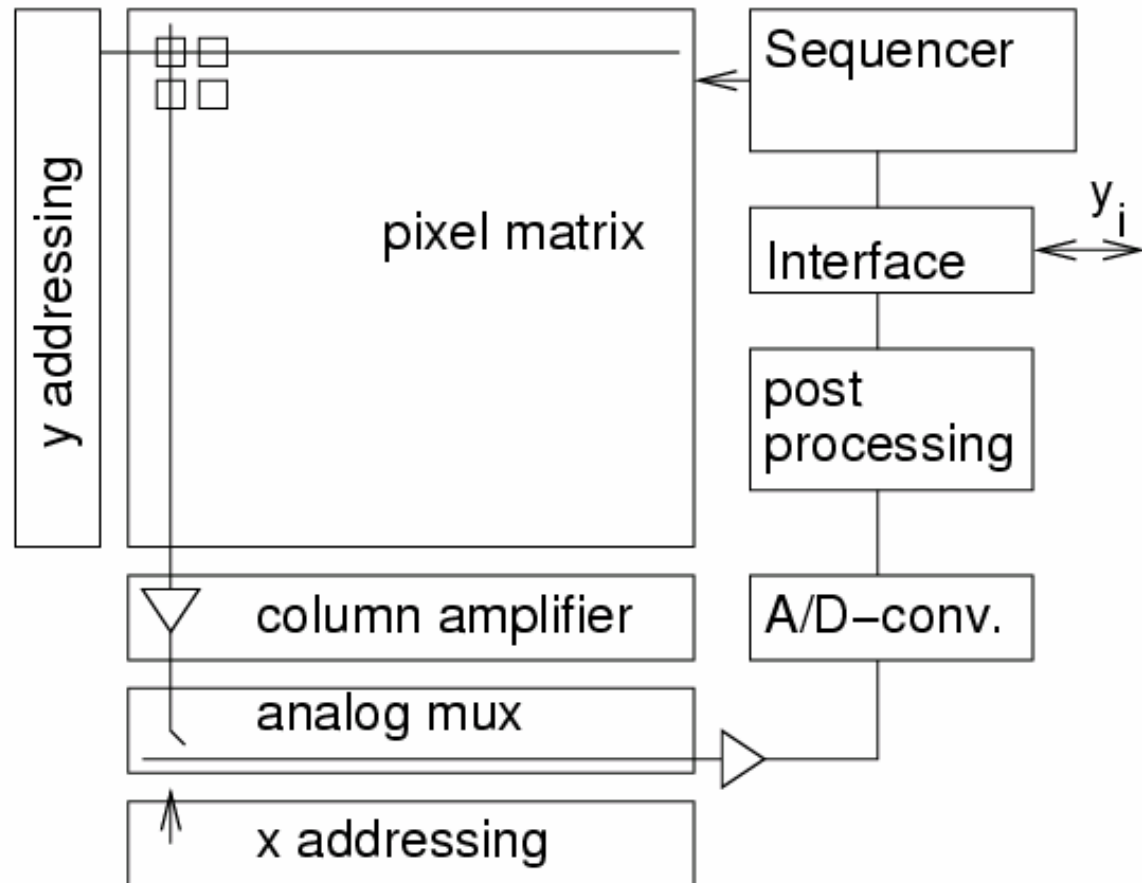
Based on charge transfer to next pixel cell



Corresponding to "bucket brigade device"
(German: "*Eimerkettenschaltung*")

CMOS image sensors

Based on standard production process for CMOS chips, allows integration with other components.



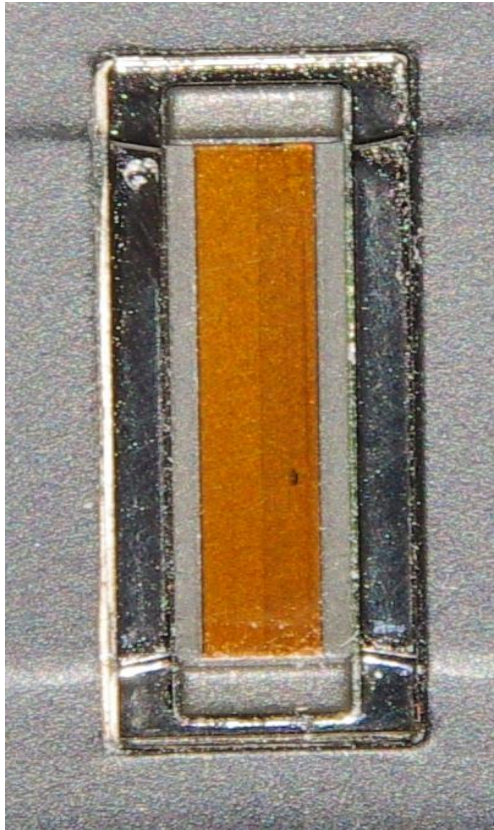
Comparison CCD/CMOS sensors

| Property | CCD | CMOS |
|---------------------------------|--------------------------------------|------------------------|
| Technology optimized for | Optics | VLSI technology |
| Technology | Special | Standard |
| Smart sensors | No, no logic on chip | Logic elements on chip |
| Access | Serial | Random |
| Size | Limited | Can be large |
| Power consumption | Low | Larger |
| Video mode | Possibly too slow | ok |
| Applications | Situation is changing over the years | |

See also B. Diericks: CMOS image sensor concepts.
Photonics West 2000 Short course (Web)

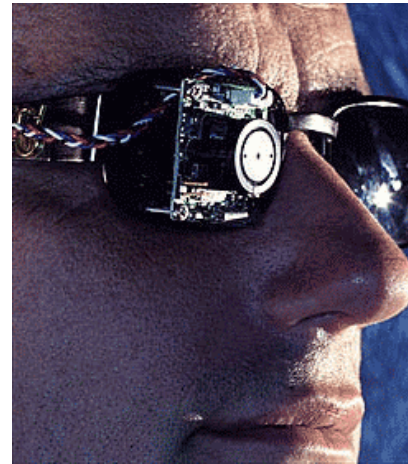
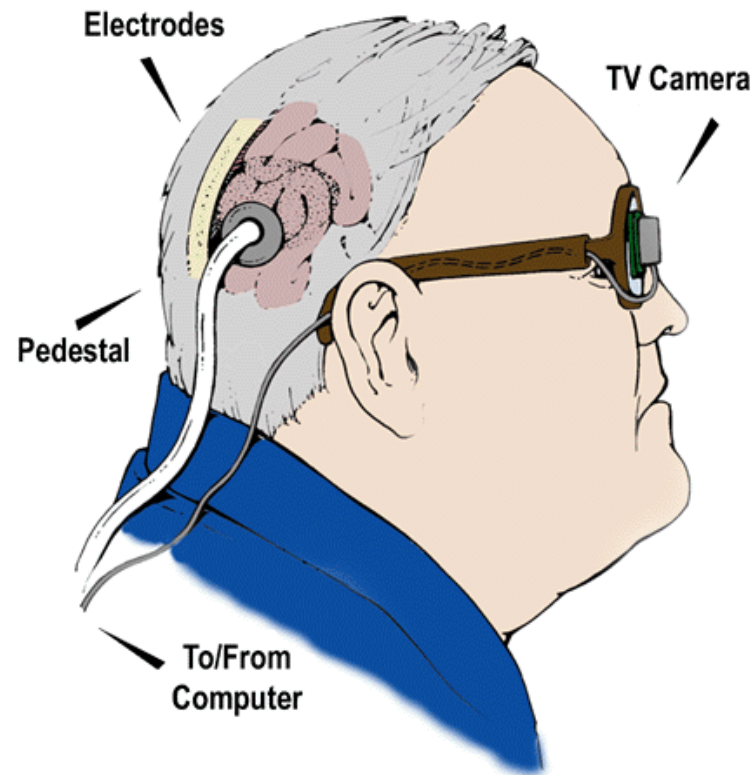
Example: Biometrical Sensors

e.g.: Fingerprint sensor



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Artificial eyes (1)



© Dobelle Institute
(was at www.dobelle.com)

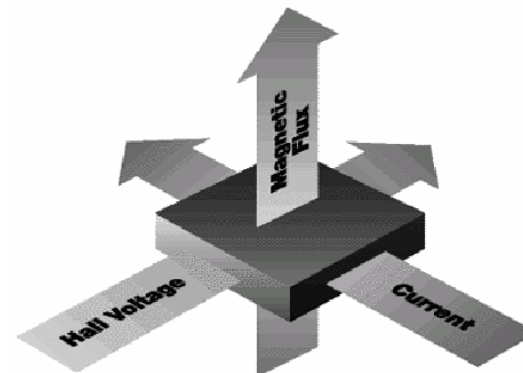
Artificial eyes (2)

- Translation into sound
[<http://www.seeingwithsound.com/etumble.htm>]



Other sensors

- Rain sensors for wiper control
(“Sensors multiply like rabbits“ [ITT automotive])
- Pressure sensors
- Proximity sensors
- Engine control sensors
- Hall effect sensors



Signals

Sensors generate *signals*

Definition: a **signal** s is a mapping

from the time domain D_T to a value domain D_V :

$$s : D_T \rightarrow D_V$$

D_T : continuous or discrete time domain

D_V : continuous or discrete value domain.

Discretization

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Discretization of time

Digital computers require discrete sequences of physical values

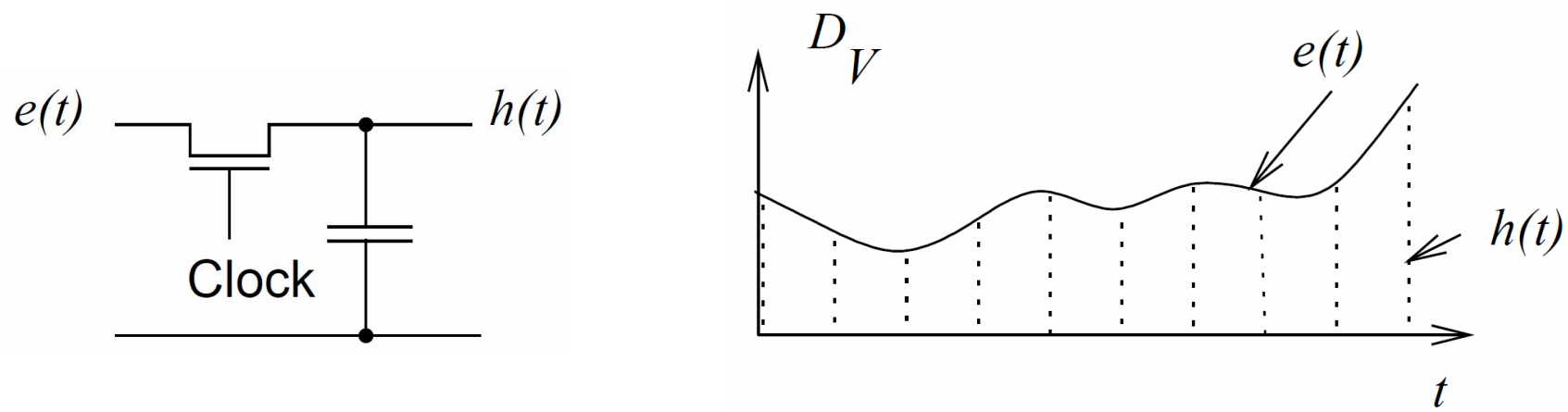
$$s : D_T \rightarrow D_V$$


Discrete time domain

☞ Sample-and-hold circuits

Sample-and-hold circuits

Clocked transistor + capacitor;
Capacitor stores sequence values



$e(t)$ is a mapping $\mathbb{R} \rightarrow \mathbb{R}$

$h(t)$ is a **sequence** of values or a mapping $\mathbb{Z} \rightarrow \mathbb{R}$

Do we lose information due to sampling?

Would we be able to reconstruct input signals from the sampled signals?

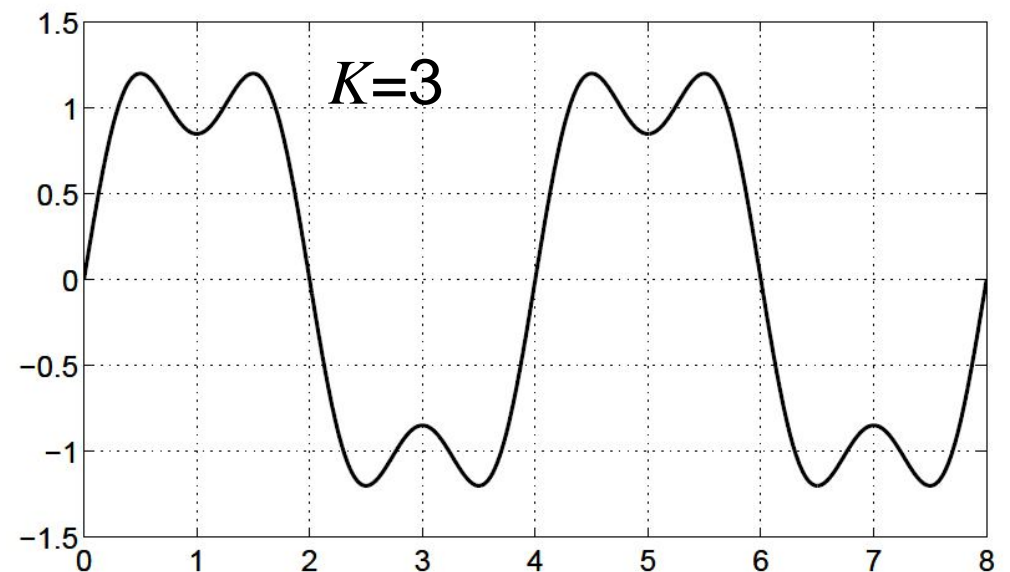
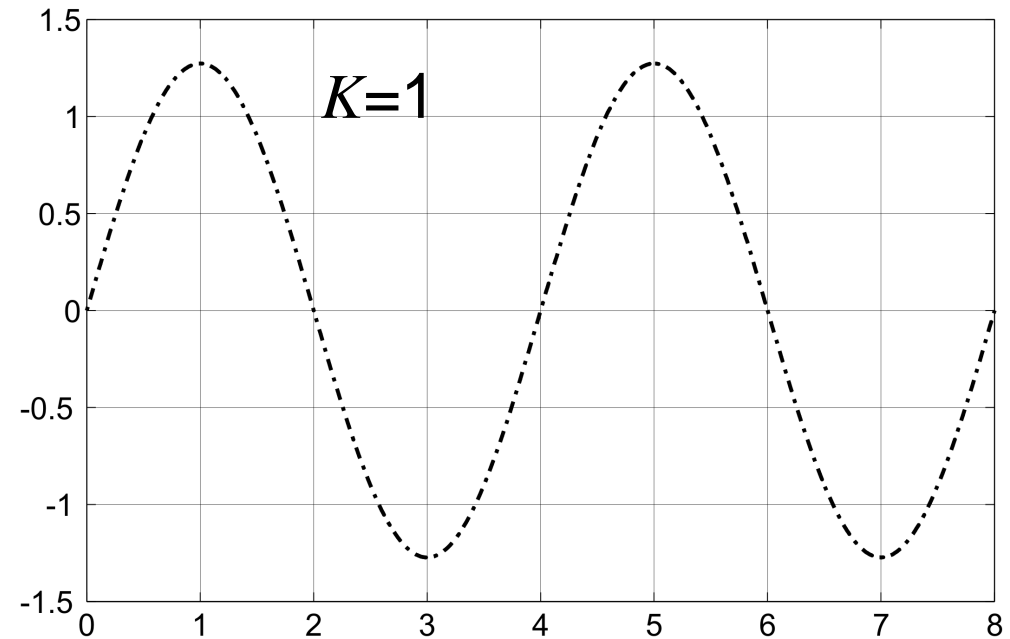
☞ approximation of signals by sine waves.

Approximation of a square wave (1)

Target: square wave
with period $p_1=4$

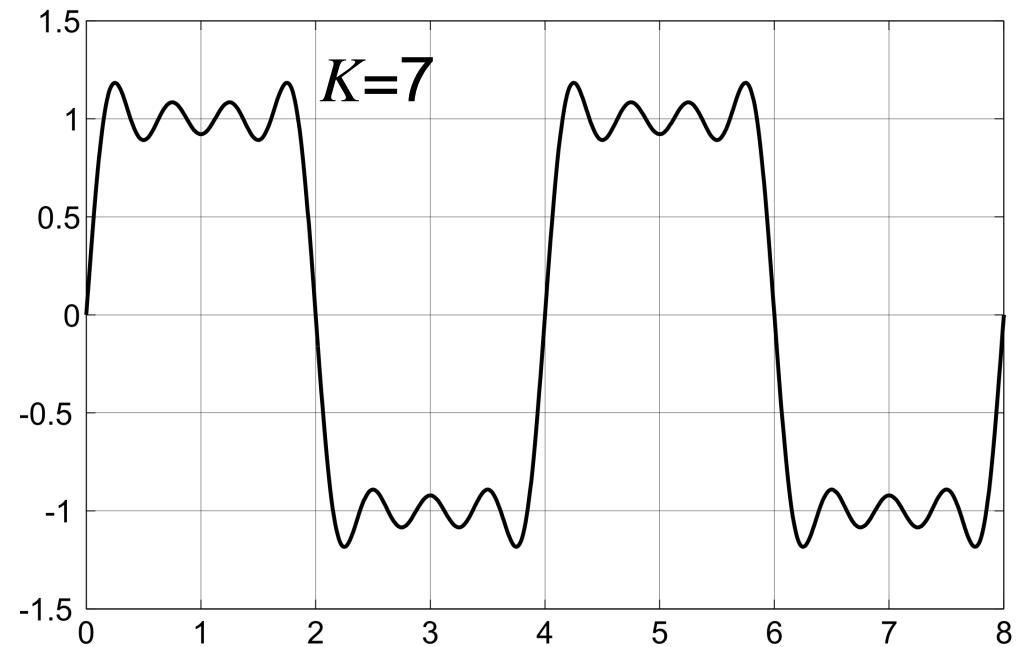
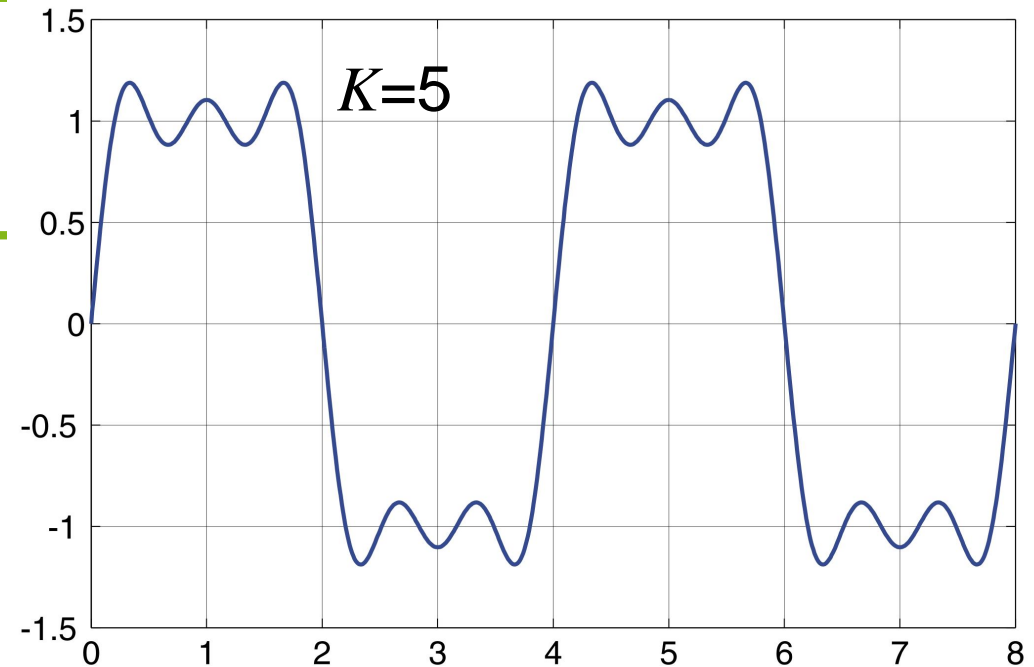
$$e'_K(t) = \sum_{k=1,3,5,\dots}^K \frac{4}{\pi k} \sin\left(\frac{2\pi t}{p_k}\right)$$

with $\forall k: p_k = p_1/k$: periods
of contributions to e'



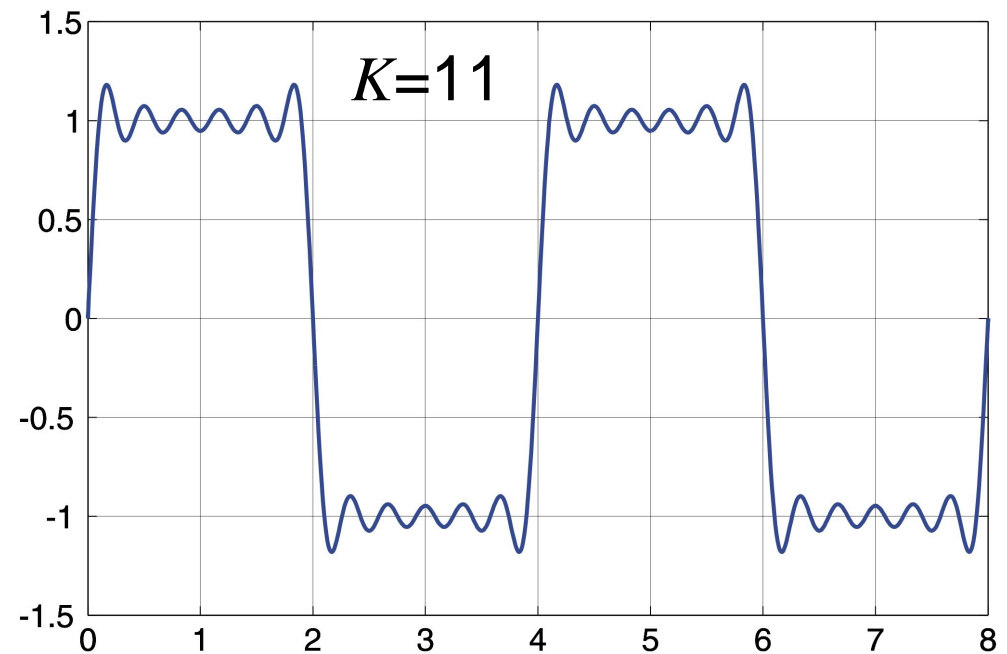
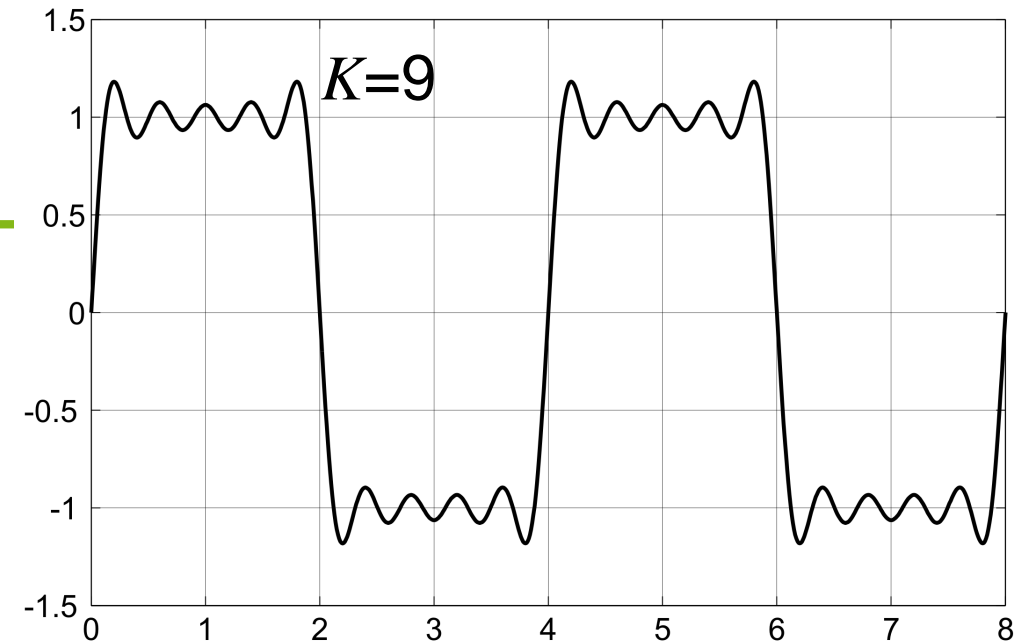
Approximation of a square wave (2)

$$e'_K(t) = \sum_{k=1,3,5,\dots}^K \frac{4}{\pi k} \sin\left(\frac{2\pi t}{4/k}\right)$$



Approximation of a square wave (3)

$$e'_K(t) = \sum_{k=1,3,5,\dots}^K \frac{4}{\pi k} \sin\left(\frac{2\pi t}{4/k}\right)$$



Linear transformations

Let $e_1(t)$ and $e_2(t)$ be signals

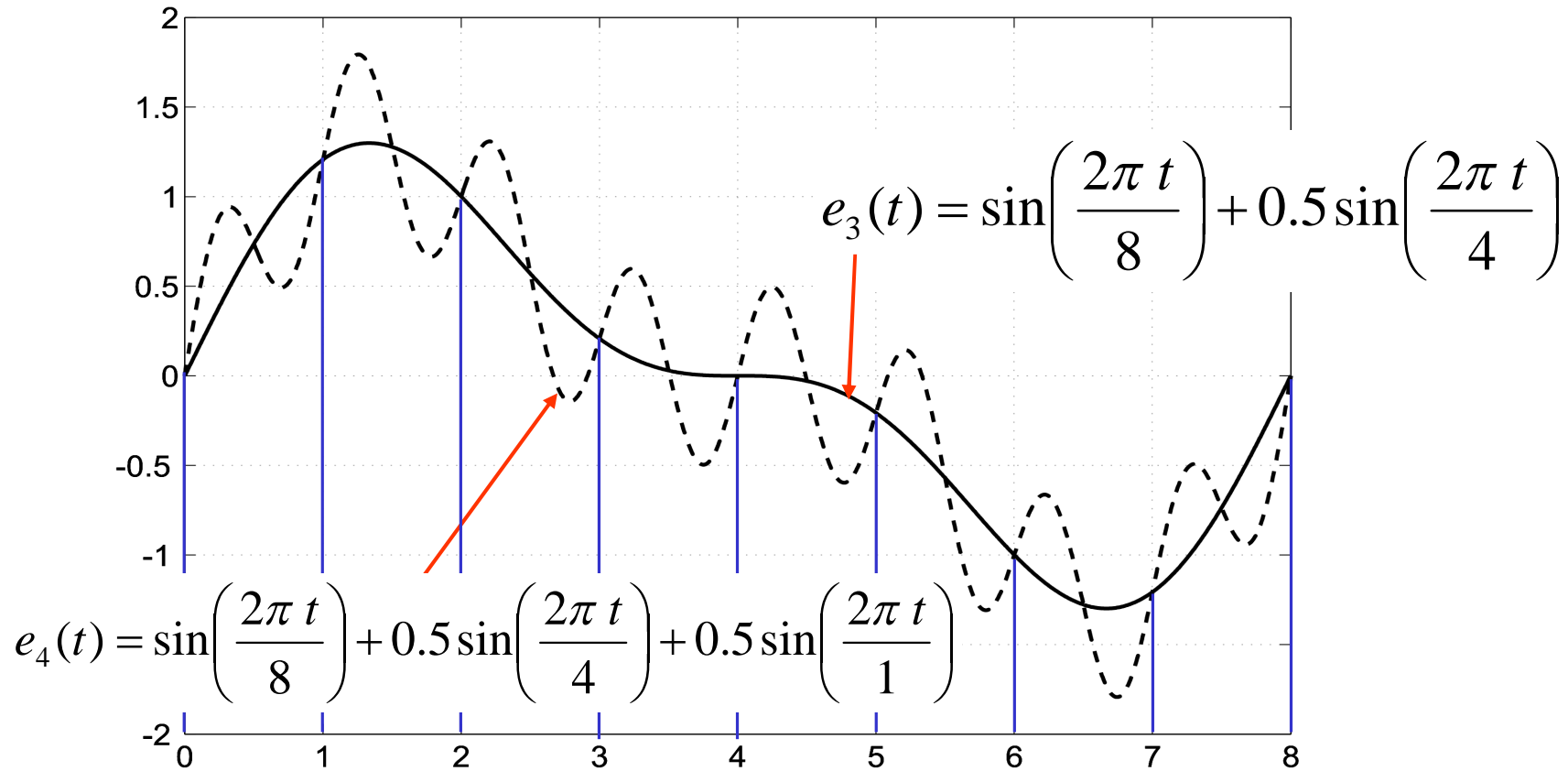
Definition: A transformation Tr of signals is linear iff

$$Tr(e_1 + e_2) = Tr(e_1) + Tr(e_2)$$

In the following, we will consider linear transformations.

☞ We consider sums of sine waves instead of the original signals.

Aliasing



Periods of $p=8,4,1$

Indistinguishable if sampled at integer times, $p_s=1$

Aliasing (2)

☞ Reconstruction impossible, if not sampling frequently enough

How frequently do we have to sample?

Nyquist criterion (sampling theory):

Aliasing can be avoided if we restrict the frequencies of the incoming signal to less than half of the sampling rate.

$p_s < \frac{1}{2} p_N$ where p_N is the period of the “fastest” sine wave

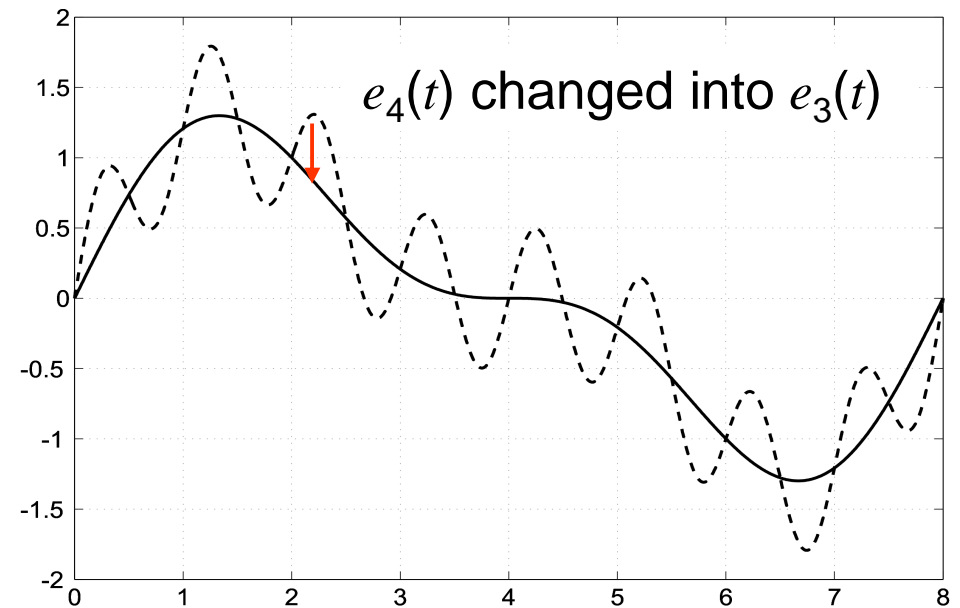
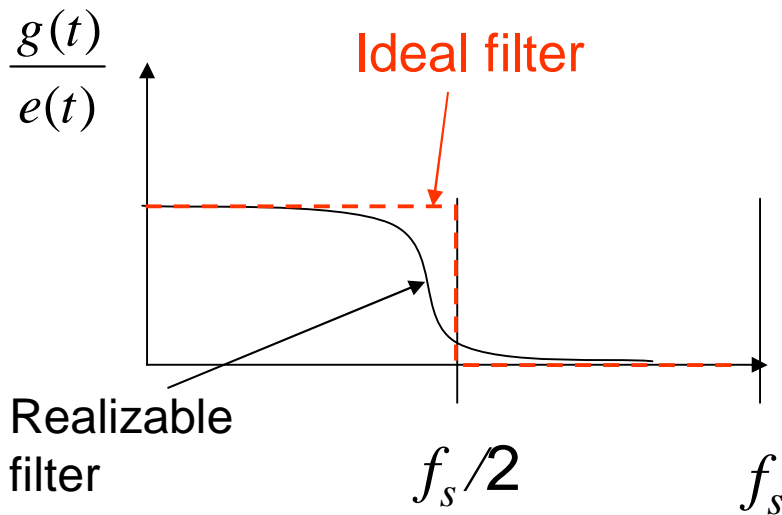
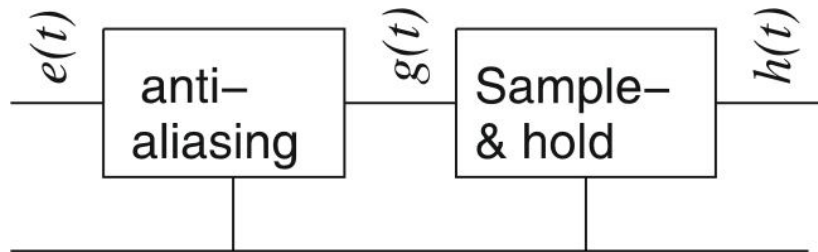
or $f_s > 2 f_N$ where f_N is the frequency of the “fastest” sine wave

f_N is called the **Nyquist frequency**, f_s is the **sampling rate**.

See e.g. [Oppenheim/Schafer, 2009]

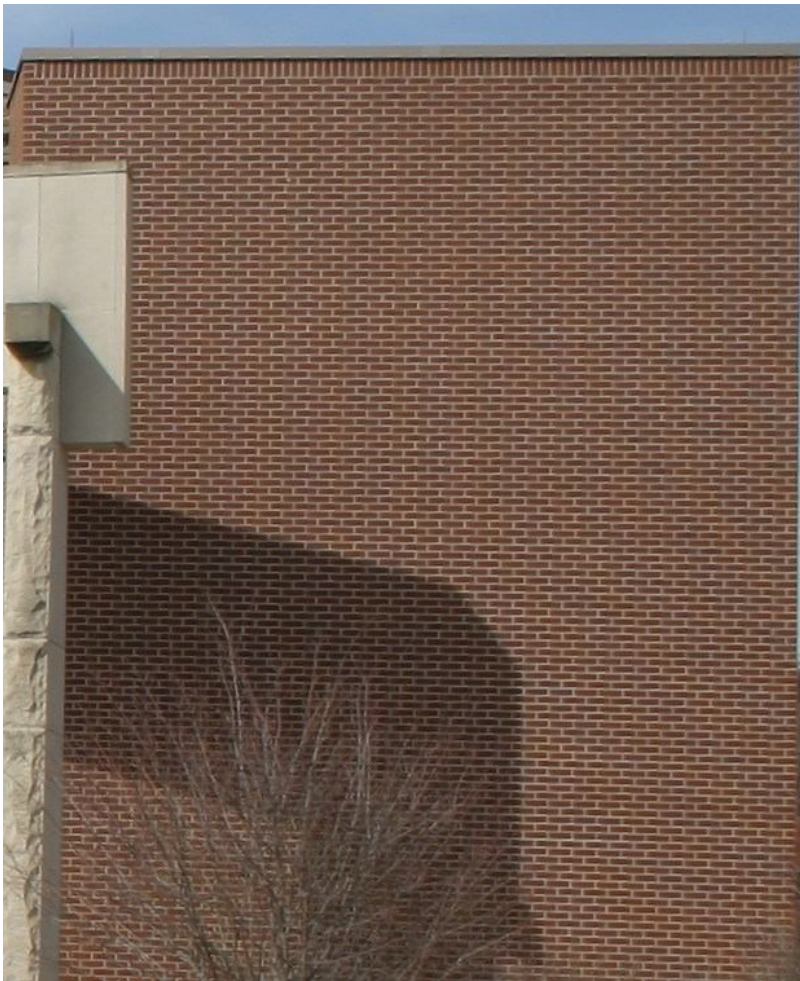
Anti-aliasing filter

A filter is needed to remove high frequencies



Examples of aliasing in computer graphics

Original

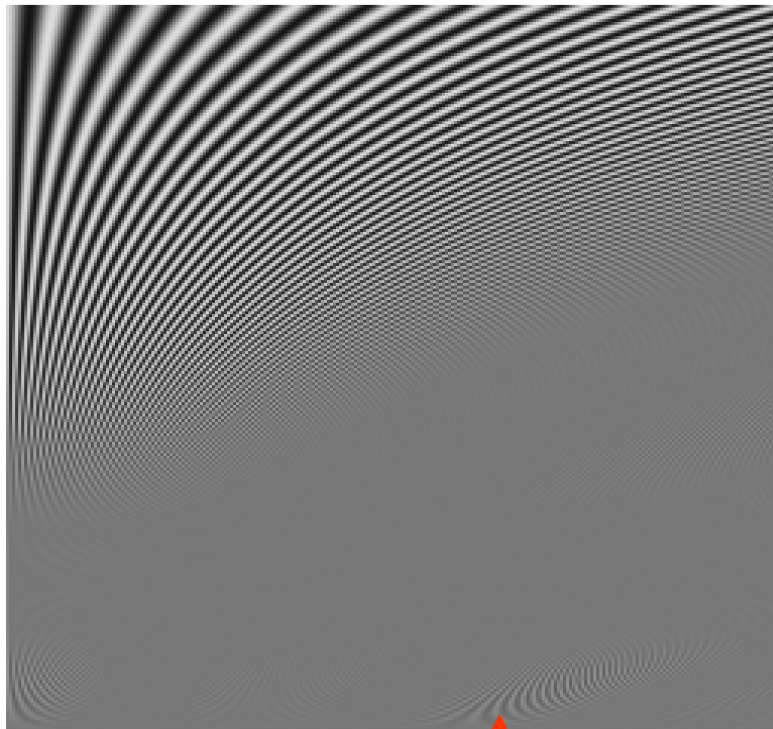


Sub-sampled, no filtering



Examples of aliasing in computer graphics (2)

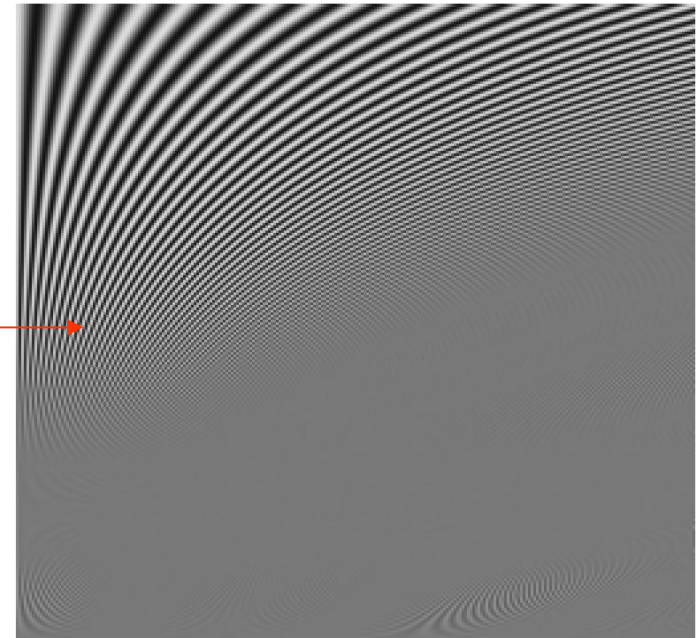
Original (pdf screen copy)



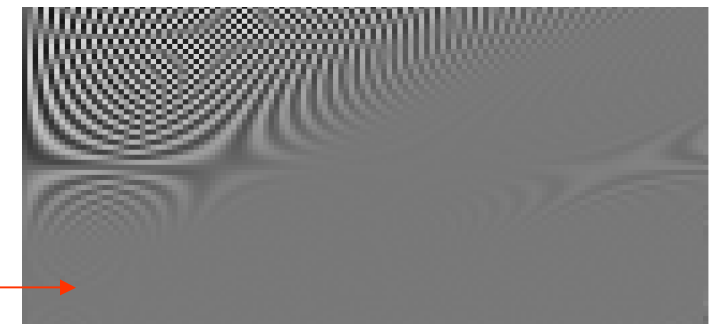
<http://www.niirs10.com/Resources/Reference Documents/Accuracy in Digital Image Processing.pdf>

Impact of rasterization

Filtered & sub-sampled



Sub-sampled, no filtering



Discretization of values: A/D-converters

Digital computers require digital form of physical values

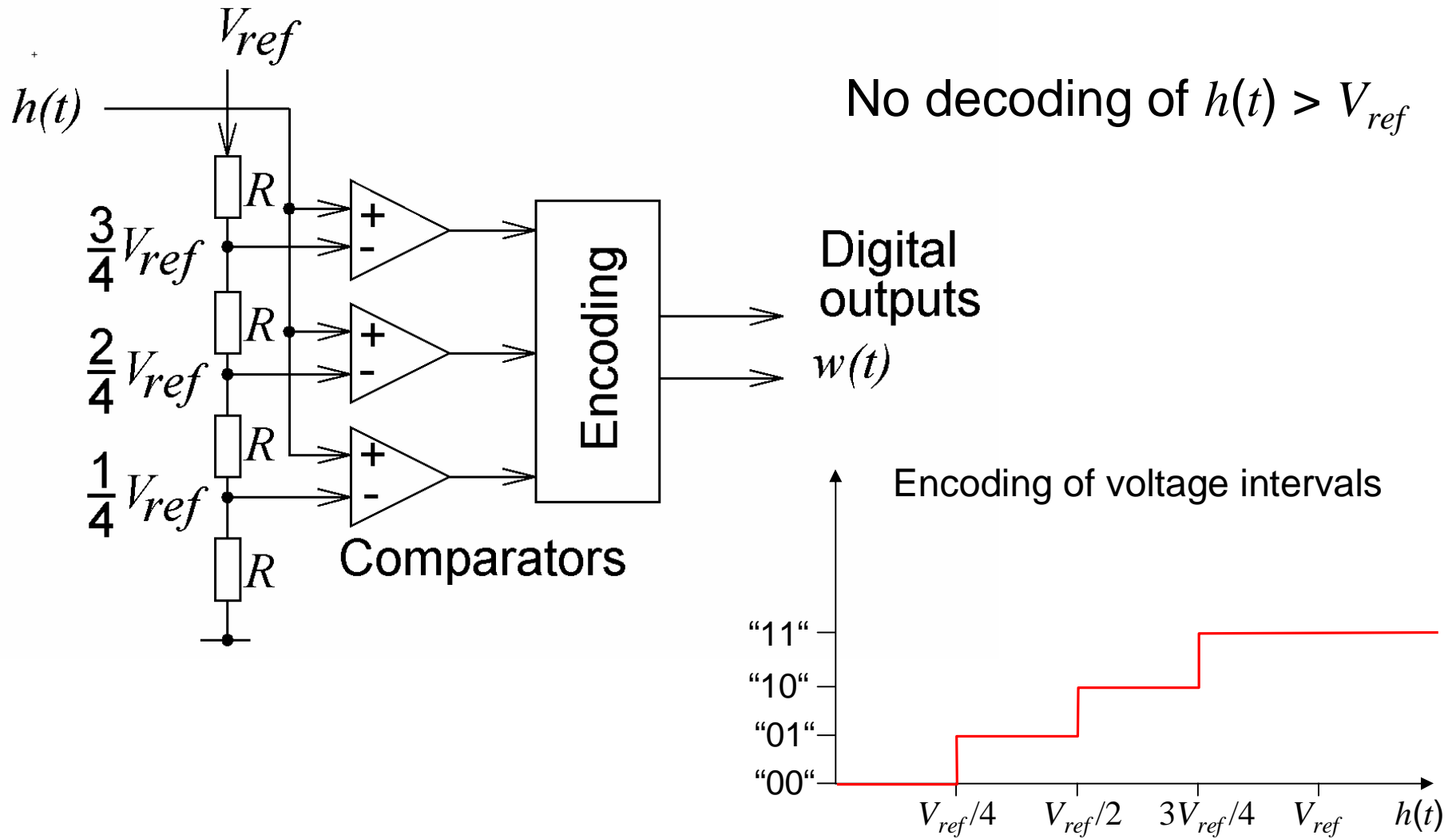
$$s: D_T \rightarrow D_V$$



Discrete value domain

☞ A/D-conversion; many methods with different speeds.

Flash A/D converter



Resolution

- Resolution (in bits): number of bits produced
- Resolution Q (in volts): difference between two input voltages causing the output to be incremented by 1

$$Q = \frac{V_{FSR}}{n} \quad \text{with}$$

Q : resolution in volts per step
 V_{FSR} : difference between largest and smallest voltage
 n : number of voltage intervals

Example:
 $Q = V_{ref}/4$ for the previous slide

Resolution and speed of Flash A/D-converter

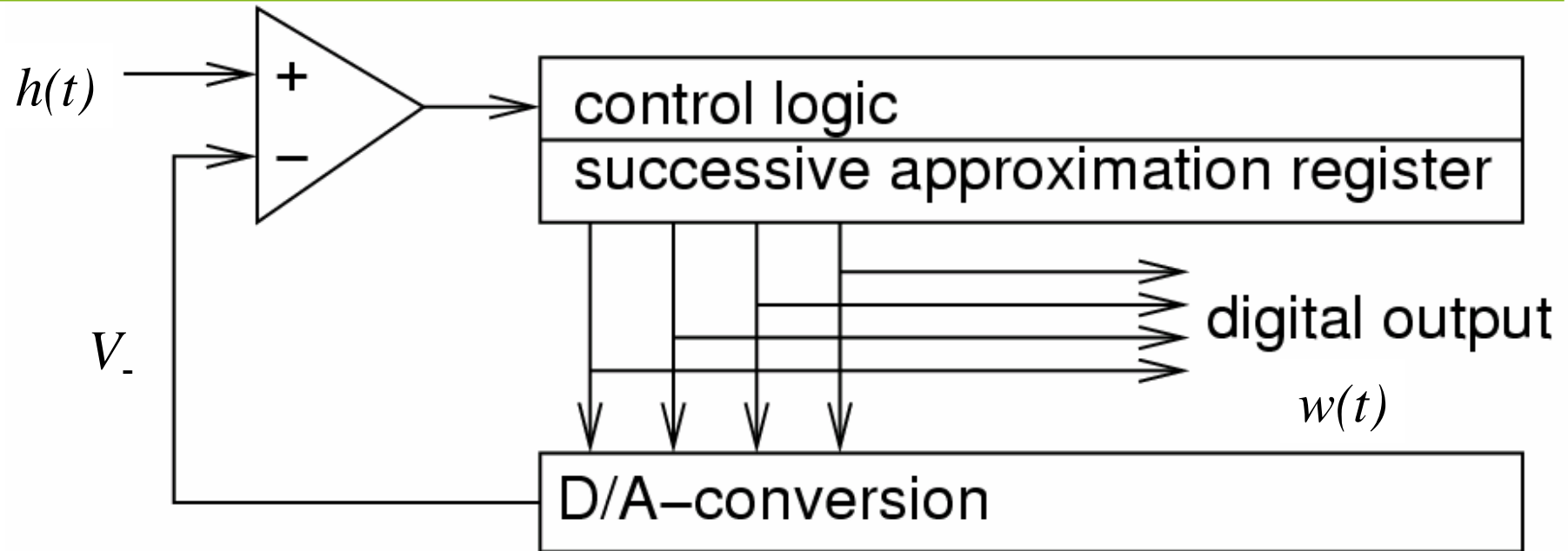
Parallel comparison with reference voltage

Speed: $O(1)$

Hardware complexity: $O(n)$

Applications: e.g. in video processing

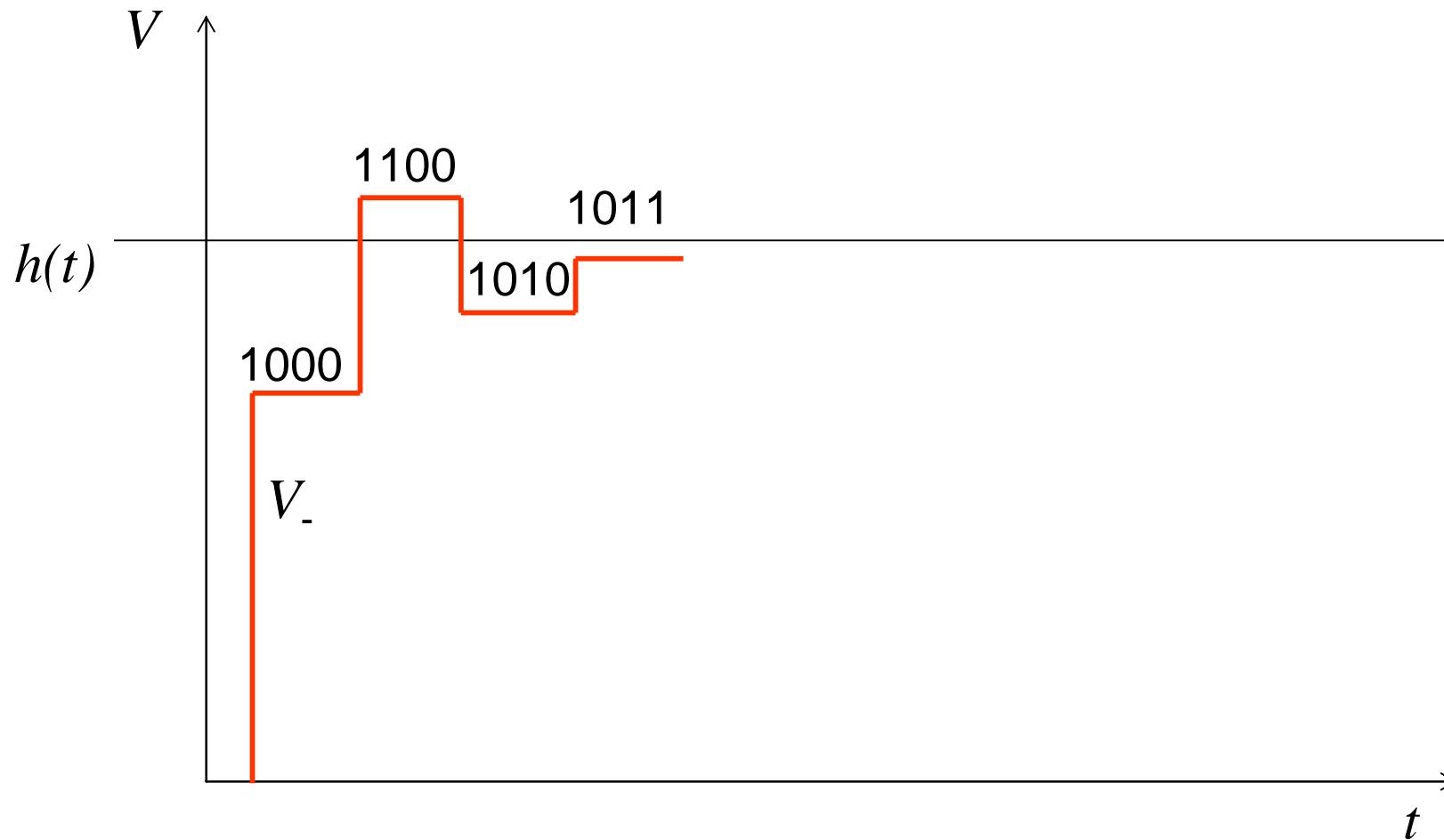
Higher resolution: Successive approximation



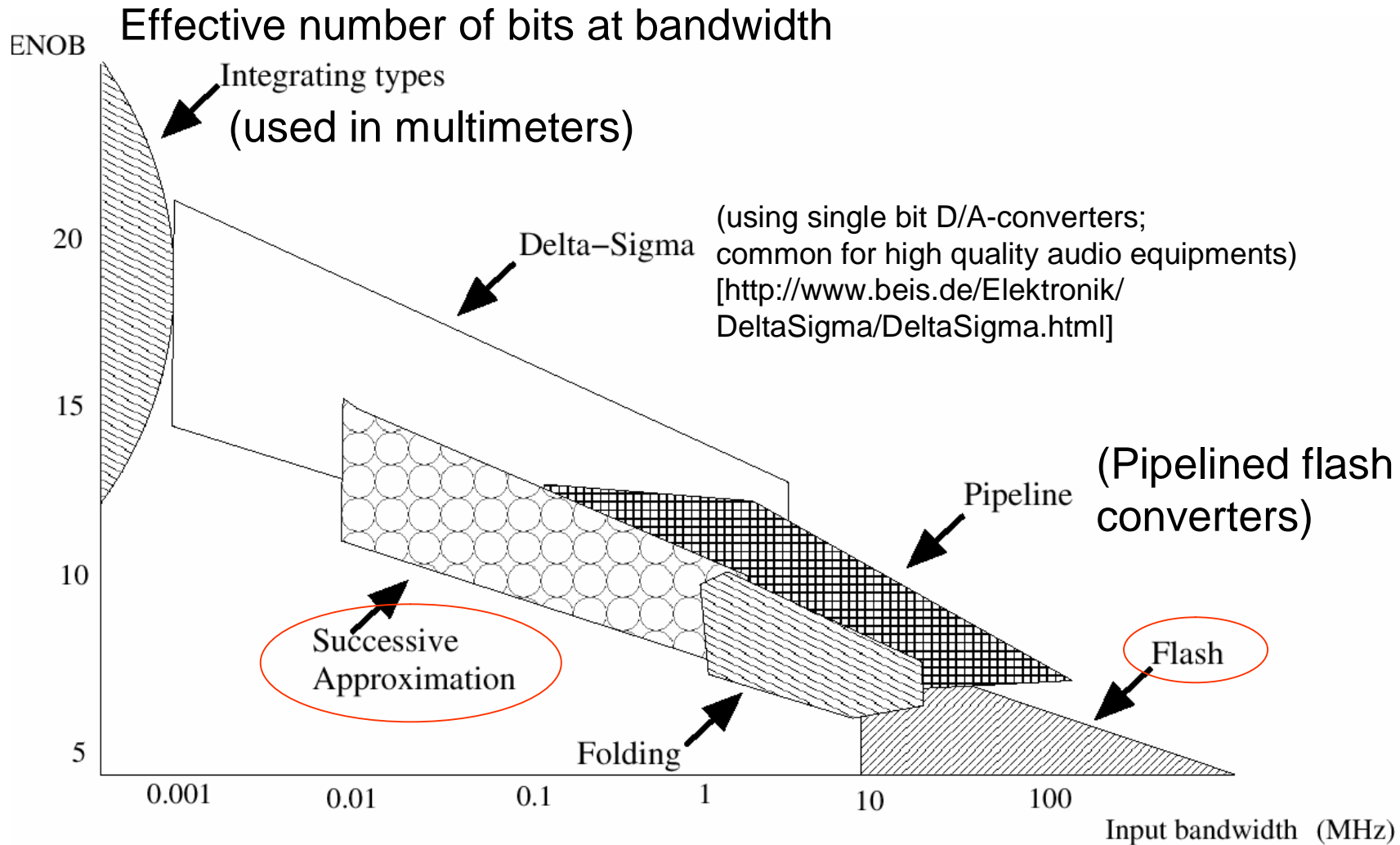
Key idea: binary search:
Set MSB='1'
if too large: reset MSB
Set MSB-1='1'
if too large: reset MSB-1

Speed: $O(\log_2(n))$
Hardware complexity: $O(\log_2(n))$
with $n = \#$ of distinguished
voltage levels;
slow, but high precision possible.

Successive approximation (2)

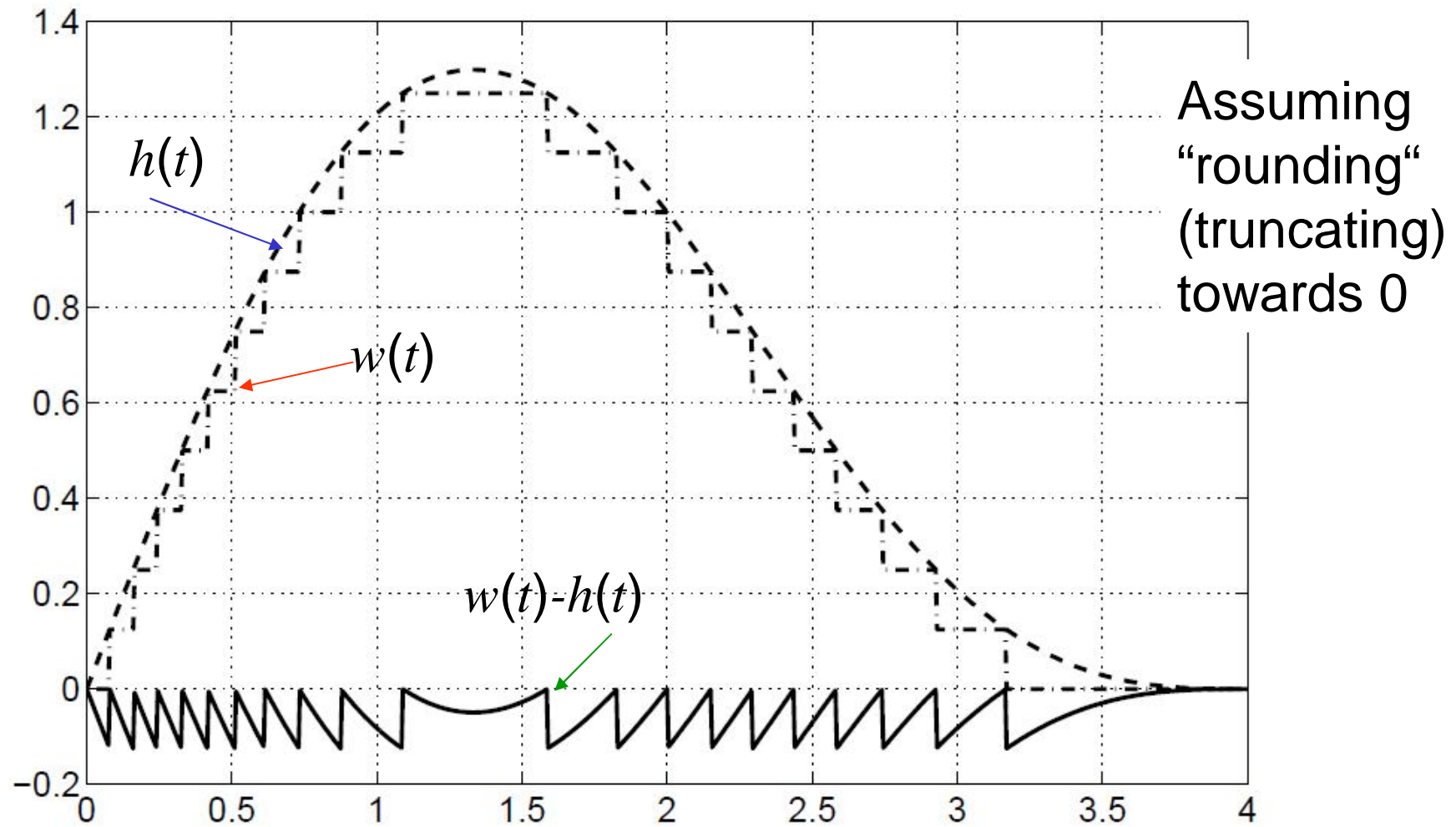


Application areas for flash and successive approximation converters

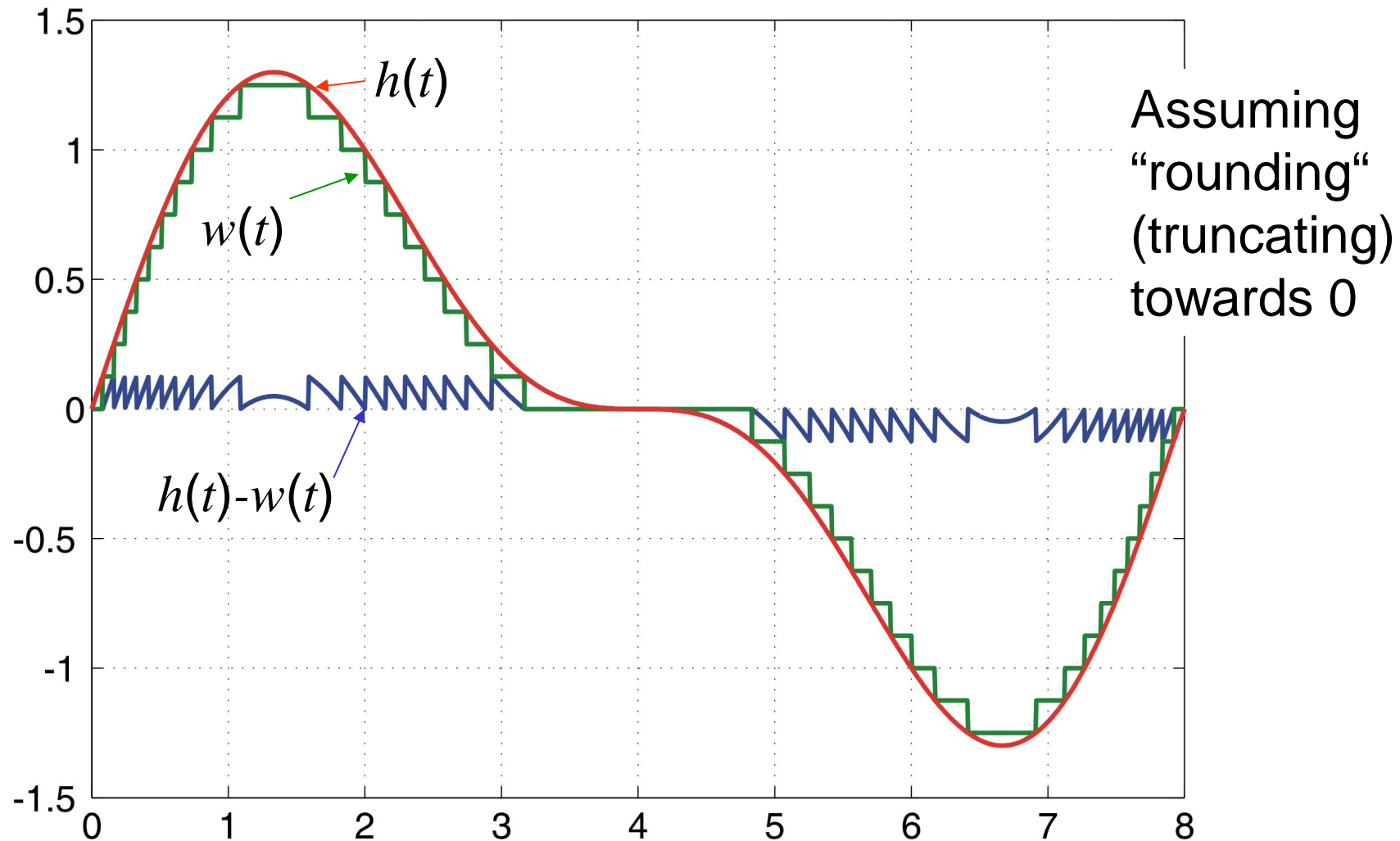


[Gielen et al., DAC 2003]

Quantization Noise



Quantization Noise



Signal to noise ratio

$$\text{signal to noise ratio (SNR) [db]} = 20 \log_{10} \left(\frac{\text{effective signal voltage}}{\text{effective noise voltage}} \right)$$

e.g.: $20 \log_{10}(2) = 6.02$ decibels

Signal to noise for ideal n -bit converter : $n * 6.02 + 1.76$ [dB]
e.g. 98.1 db for 16-bit converter, ~ 160 db for 24-bit converter

Additional noise for non-ideal converters

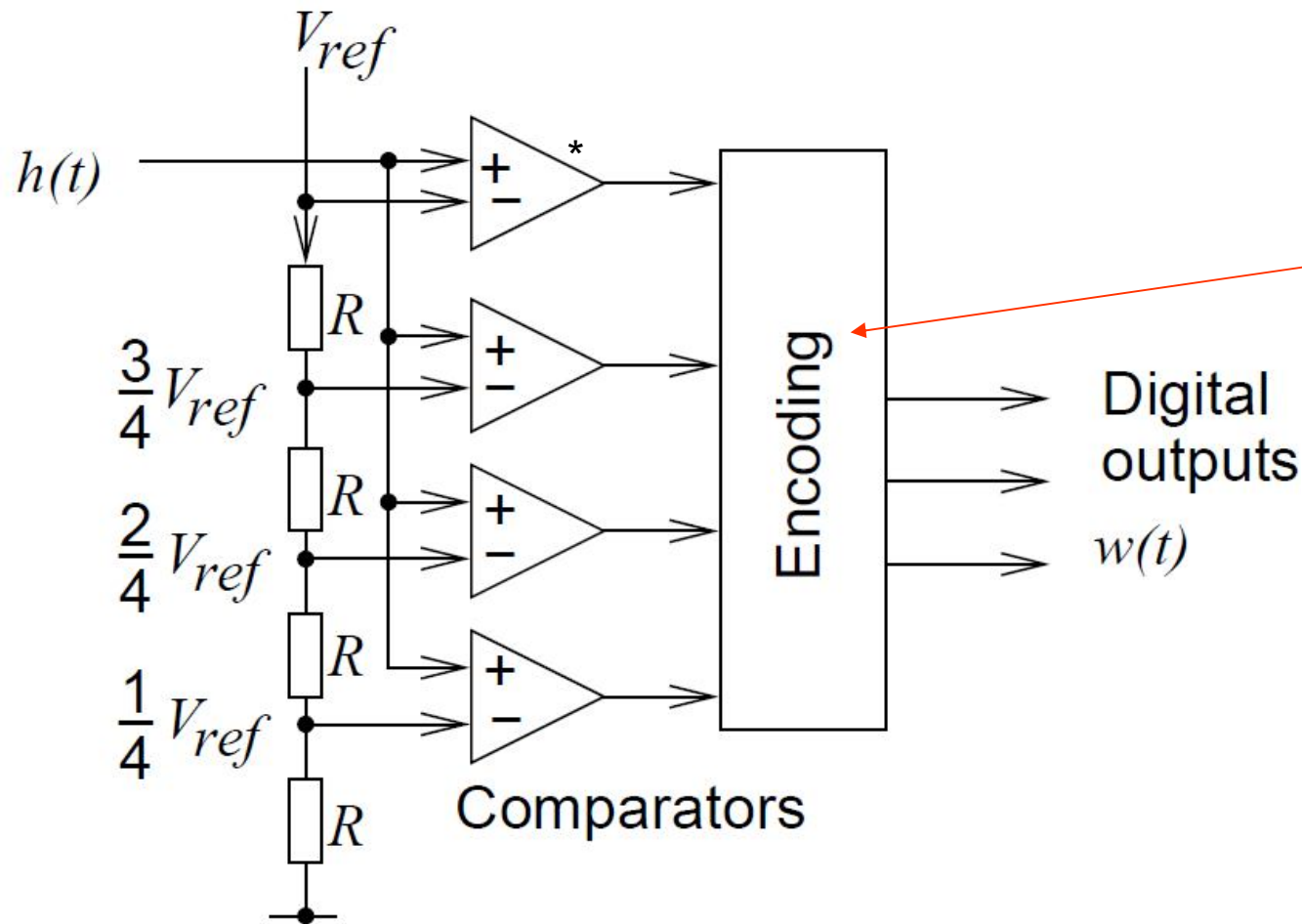
Summary

Hardware in a loop

- Sensors
- Discretization
 - Sample-and-hold circuits
 - Aliasing (and how to avoid it)
 - Nyquist criterion
 - A/D-converters
 - Quantization noise

SPARE SLIDES

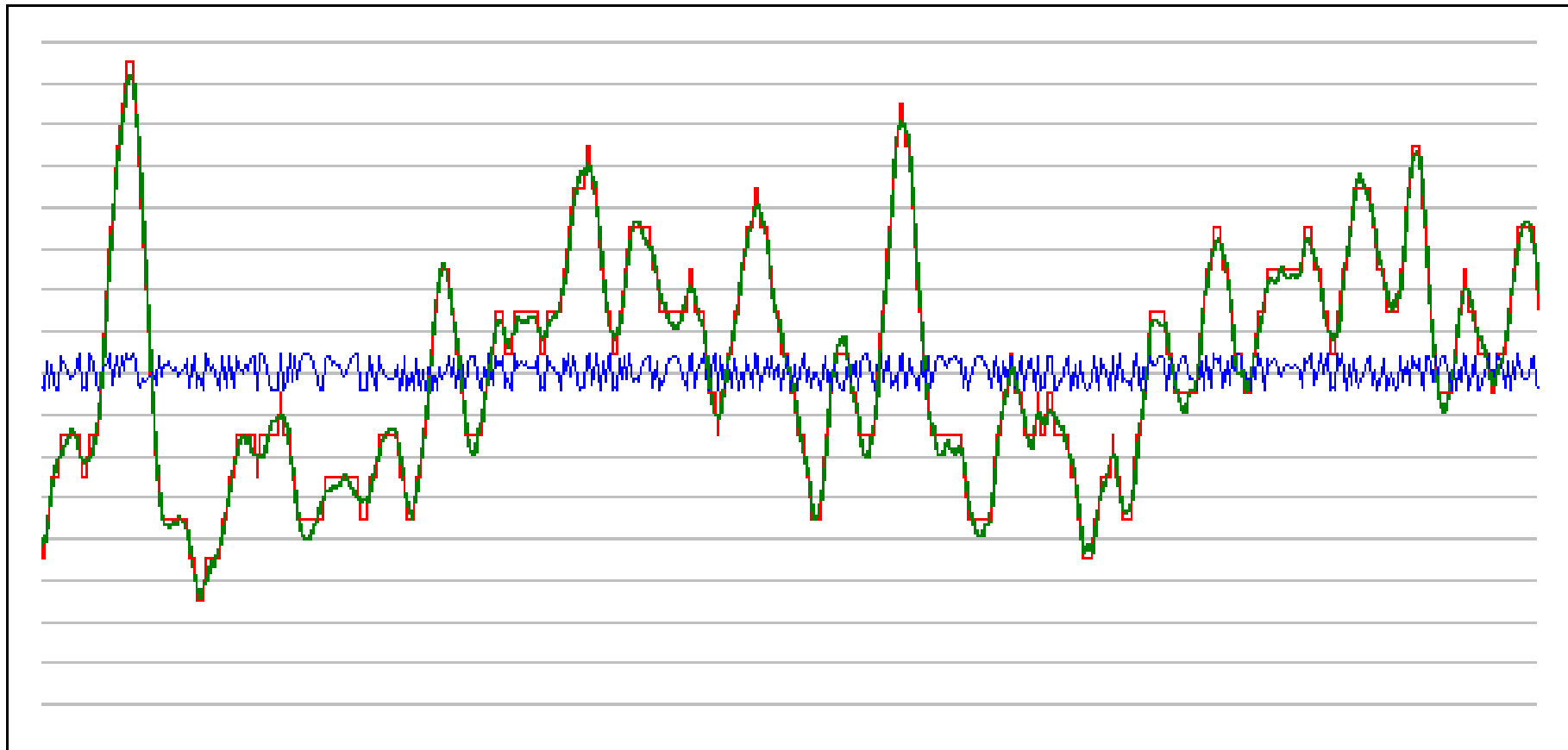
Flash A/D converter



Encodes input number of most significant '1' as an unsigned number, e.g.
 "1111" -> "100",
 "0111" -> "011",
 "0011" -> "010",
 "0001" -> "001",
 "0000" -> "000"
 (Priority encoder).

* Frequently, the case $h(t) > V_{ref}$ would not be decoded

Quantization noise for audio signal



Source: [<http://www.beis.de/Elektronik/DeltaSigma/DeltaSigma.html>]