

# **EDISP**

## **(English) Digital Signal Processing**

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## General information

- Lectures** 2h/week, Tue, 14:15-16 → *Mind the exceptions – see the schedule!*
- Labs** ≈4h/2weeks: Monday 8:15-12, room 022. See the schedule. *First meeting for all students – Monday 8 Oct, 9:15*
- Contact** J. Misiurewicz, (jmisiure@elka.pw.edu.pl) room 447. M. Malanowski (mmalanow@ise.pw.edu.pl) room 453.
- Web page** <http://staff.elka.pw.edu.pl/~jmisiure/> → Slides on Monday evening! (usually ;-)
- Homeworks** Announced as a preparation for the tests.
- Exams** Two short tests within lecture hours (see the lab schedule) and a final exam during the winter exam session (TBA).
- Scoring:**
- |       |   |     |                                       |
|-------|---|-----|---------------------------------------|
| 2x10% | = | 20% | tests                                 |
| 6x5%  | = | 30% | lab + entry test (lab 0 – not scored) |
|       |   | 50% | final exam                            |
| 2x2%  | = | 4%  | extra for homeworks                   |
- Short path** `if [(score≥41)&&(tests≥15)&&(test2≥5)]; then score* = 2; fi`
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## Books

**base book** The course is based on selected chapters of the book:

A. V. Oppenheim, R. W. Schaffer, *Discrete-Time Signal Processing*, Prentice-Hall 1989 (or II ed, 1999; also previous editions: *Digital Signal Processing*).

**free book** A free textbook covering some of the subjects can be found here: <http://www.dspguide.com/pdfbook.htm> *The book is slightly superficial, but it can be valuable*

**Additional books** available in Poland:

R.G. Lyons, *Wprowadzenie do cyfrowego przetwarzania sygnałów* (WKiŁ 1999)

Craig Marven, Gilian Ewers, *Zarys cyfrowego przetwarzania sygnałów*, WKiŁ 1999 [en: A simple approach to digital signal processing, Wiley & Sons, 1996]

Tomasz P. Zieliński, *Od teorii do cyfrowego przetwarzania sygnałów*, WKiŁ 2002 (and next edition with slightly modified title)

You may also buy/borrow a laboratory scriptbook for a Polish language course (*Cyfrowe Przetwarzanie Sygnałów*, red. A Wojtkiewicz, Wydawnictwa PW) – but our lab is different!

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A schedule was here - see the webpage for an updated version!

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## What Is EDISP All About ;-)

**Theory** Discrete-time signal processing

**Practice** Digital signal processing

Application examples:

**Filters** Guitar effects, radar, software radio, medical devices...

**Adaptive filters** Echo canceller, noise cancellation (e.g. hands-free microphone in a car),...

**Discrete Fourier Transform/FFT** Signal analyzer, OFDM modulation, Doppler USG, ...

**Random signals** Voice compression, voice recognition....

**2D signals** Image processing, USG/tomography image reconstruction, directional receivers,

...

**Upsampling/Interpolation** CD audio output, ....

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## Oversampling CD audio D/A conversion (example)

Please have a look at the black/green-board.

Notice & remember some things:

- Upsampling
  - Filtering (and what happens to the signal spectrum)
  - Frequency response (frequency characteristics) of a filter
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## Signal classification

Continuous or Discrete **amplitude** and **time**.

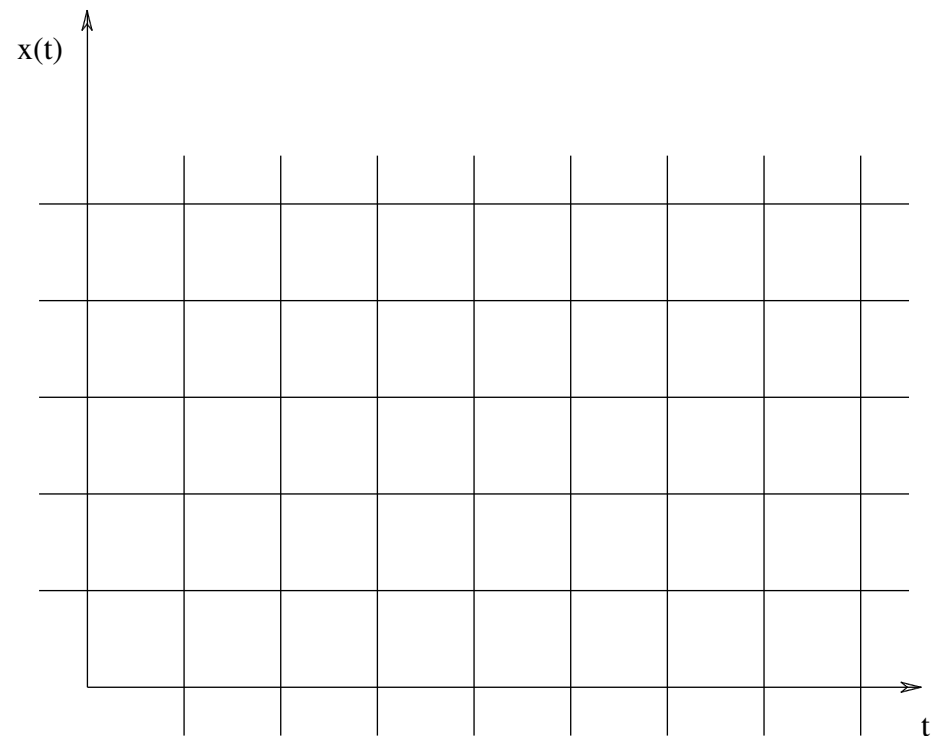
**CA-CT** → “analog” signals

**DA-CT** →

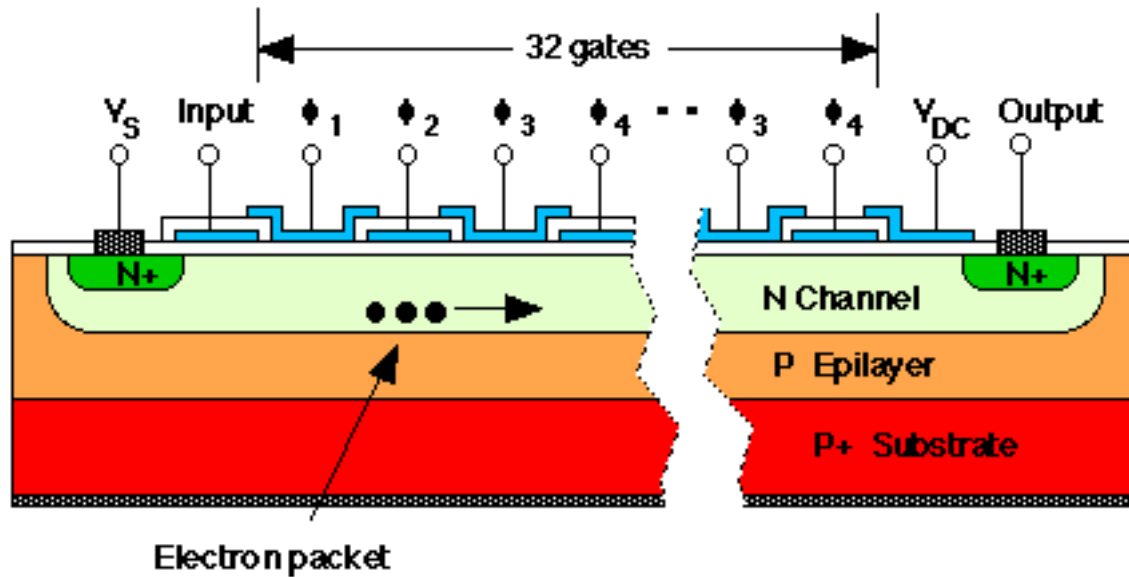
**CA-DT** → CCD, SC, SAW devices

**DA-DT** → digital devices

We’ll speak mainly about DT properties; only in some subject DA will be of importance.



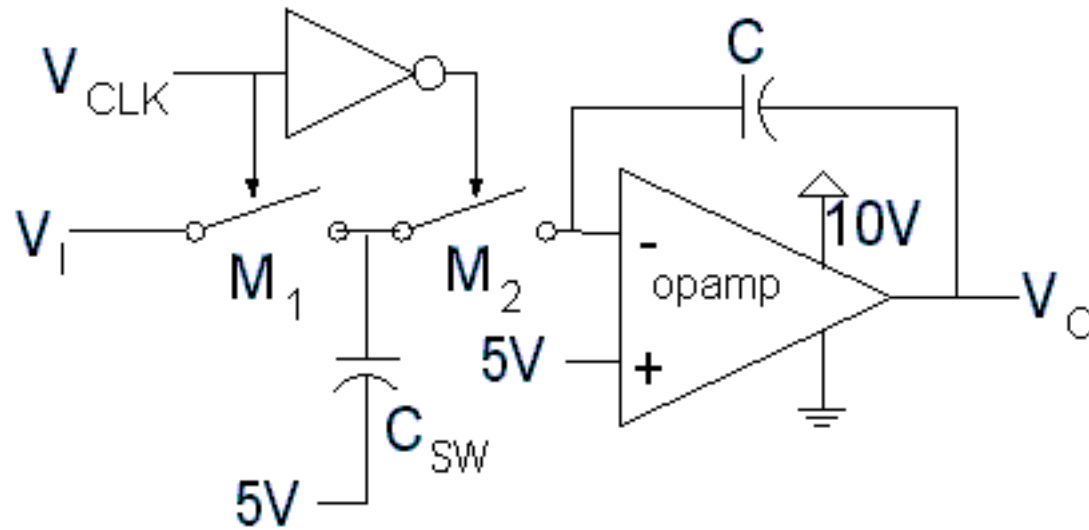
## CCD device (side remarks)



Charge is transferred on the clock edge (discrete time!).  
Clock is usually polyphase (2-4 phases).



### SC device (side remarks)



## DT signal representations

DT signal  $\longleftrightarrow$  a number sequence

$$x[n] = \{x(n)\}$$

$x[n]$  is a number sequence (or ...)

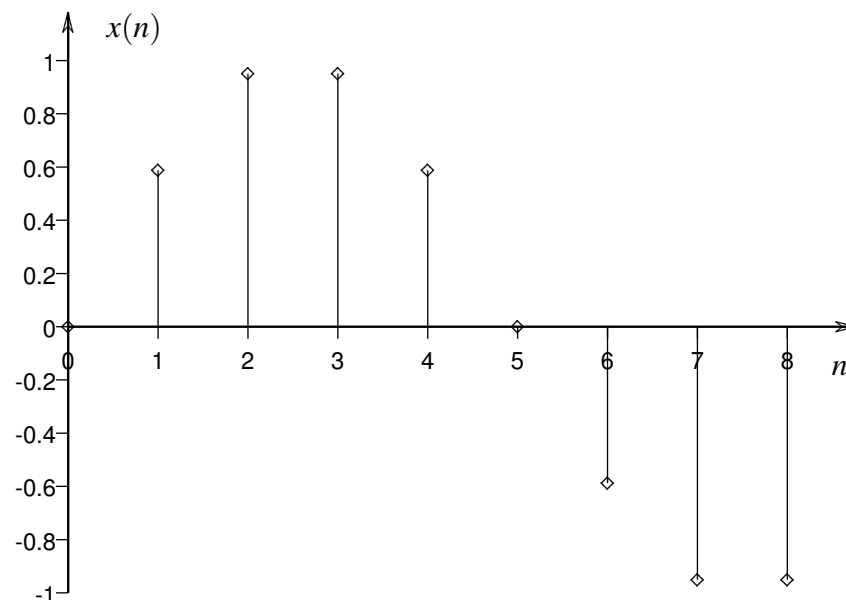
$x(n)$  is a  $n$ -th sample

$\longrightarrow$   $x(n)$  is *undefined* for  $n \notin I$

- it *may* come from sampling of analog signal
- but it may also be inherently discrete
- $n$  may correspond to: time, space,

...

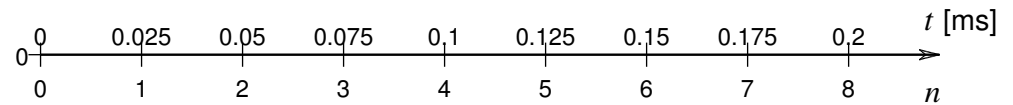
However, the most popular interpretation is: periodic sampling in time.



## Periodic sampling

$$n \longleftarrow \longrightarrow n \cdot T_s$$

$$x(n) = x_a(nT_s)$$



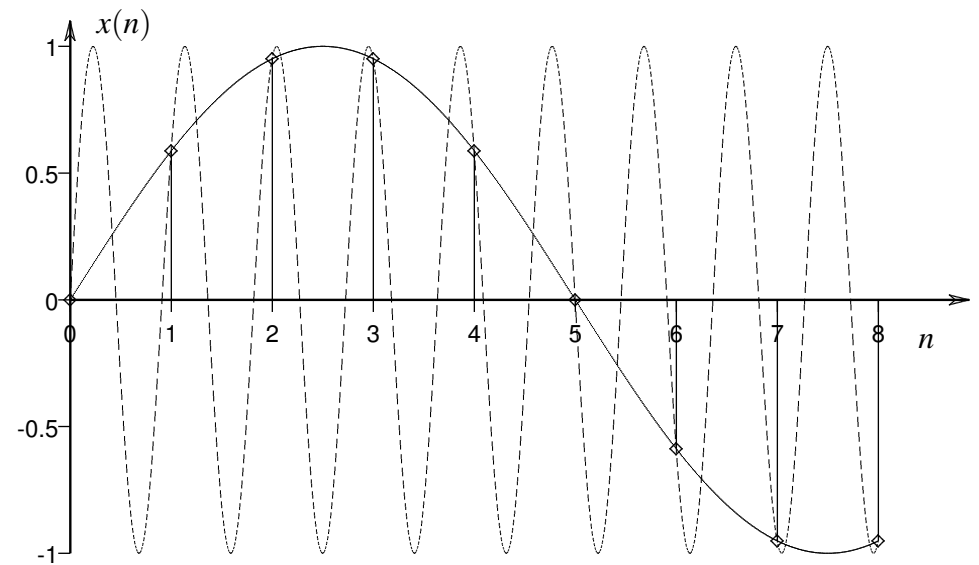
$$n = t/T_s, \quad T_s = 0.025 \text{ [ms]}$$

### Misinterpretations

→ we do not know what is between points

a)  $\sin(n \cdot (1/5) \cdot \pi)$  or

b)  $\sin(n \cdot (2 + 1/5) \cdot \pi)$  ?



We have to **know** which one to choose → sampling theorem