EDISP (English) Digital Signal Processing

Jacek Misiurewicz *e-mail: jmisiure@elka.pw.edu.pl* Institute of Electronic Systems Warsaw University of Technology Warsaw, Poland

October 5, 2009

General information

Lectures	2h/week, Tue, 14:15-16			
Labs	pprox4h/2weeks: Monday 8:15-12, room 022. See the schedule.			
	First meeting for all students – Monday 12 Oct, 9:15			
Contact	J. Misiurewicz, (jmisiure@elka.pw.edu.pl) room 447.			
Web page	e http://staff.elka.pw.edu.pl/~jmisiure/			
	\longrightarrow 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).				
	2x10%	=	20%	tests
Scoring:	6x5%	=	30%	lab + entry test (lab 0 – not scored)
			50%	final exam
	2x2%	=	4%	extra for homeworks
Short path if $[(score \ge 41)\&\&(tests \ge 15)\&\&(test2 \ge 5)];$ then $score* = 2;$ fi				

Books

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

A. V. Oppenheim, R. W. Schafer, *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!

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

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,

Oversampling CD audio D/A conversion (example)

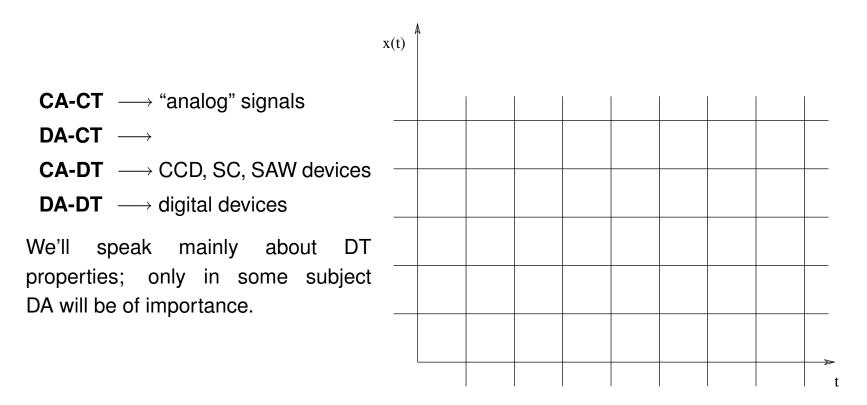
Please have a look at the black/green-board.

Notice & remember some things:

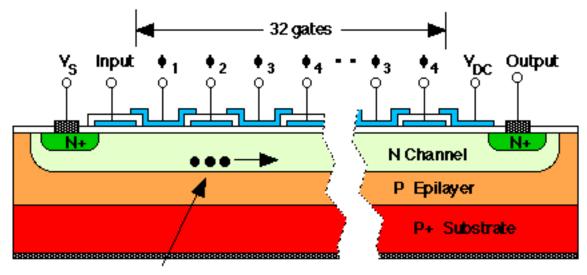
- Upsampling
- Filtering (and what happens to the signal spectrum)
- Frequency response (frequency characteristics) of a filter

Signal classification

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



CCD device (side remarks)

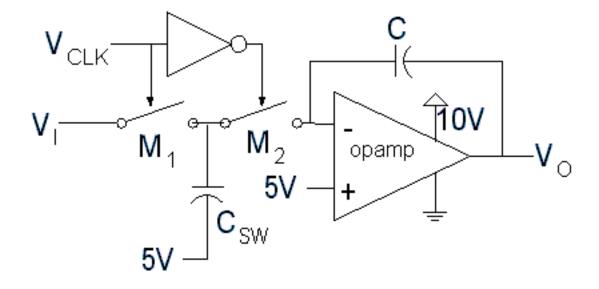


Electronpacket

Charge is transferred on the clock edge (discrete time!).

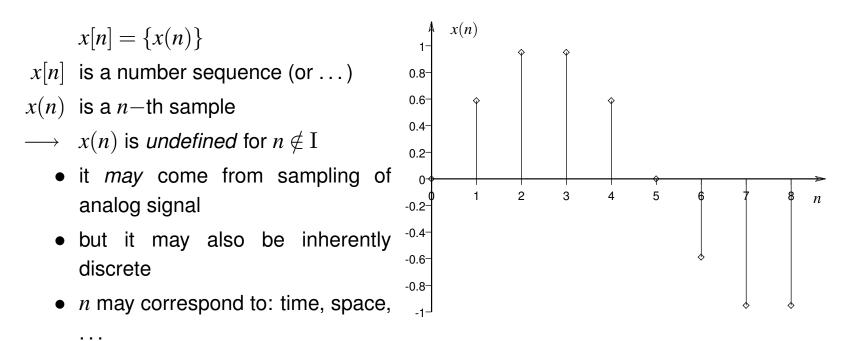
Clock is usually polyphase (2-4 phases).

SC device (side remarks)



DT signal representations

DT signal \longleftrightarrow a number sequence



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

 \rightarrow

Periodic sampling t [ms] п $n \leftarrow m \cdot T_s$ $n = t/T_s, T_s = 0.025 \text{ [ms]}$ $x(n) = x_a(nT_s)$ $1 \overset{x(n)}{\uparrow} \overset{x(n)}{\land}$ **Misinterpretations** 0.5we do not know what is between points 0-2 3 8 п a) $sin(n \cdot (1/5) \cdot \pi)$ or -0.5b) $sin(n \cdot (2+1/5) \cdot \pi)$? -1-

We have to **know** which one to choose \longrightarrow sampling theorem