Digital Signal Processing

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Digital Signal Processing

Lecture 11

Linearity & Time-Invariance Convolution

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READING ASSIGNMENTS

- This Lecture:
 - Chapter 5, Sections 5-5 and 5-6
 - Section 5-4 will be covered, but not "in depth"
- Other Reading:
 - Recitation: Ch. 5, Sects 5-6, 5-7 & 5-8
 - CONVOLUTION
 - Next Lecture: start Chapter 6

LECTURE OBJECTIVES

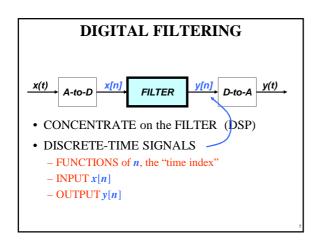
- GENERAL PROPERTIES of FILTERS
 - LINEARITY

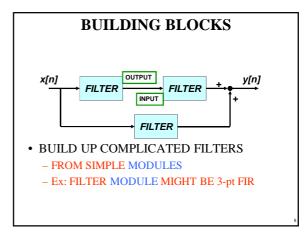
LTI SYSTEMS

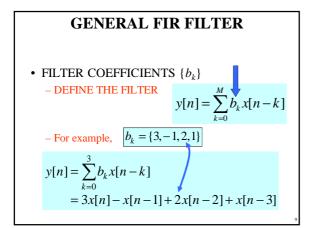
- TIME-INVARIANCE
- ==> CONVOLUTION
- BLOCK DIAGRAM REPRESENTATION
 - Components for Hardware
 - Connect Simple Filters Together to Build More Complicated Systems

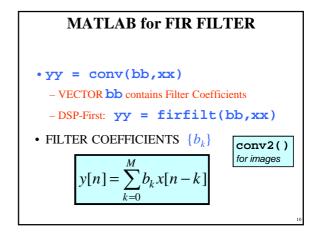
OVERVIEW

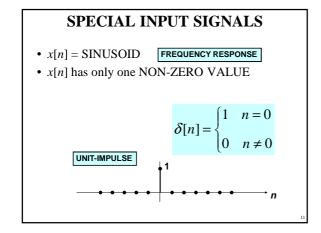
- h[n]• IMPULSE RESPONSE.
 - FIR case: same as $\{b_k\}$
- CONVOLUTION - GENERAL: y[n] = h[n] * x[n]
 - GENERAL CLASS of SYSTEMS
 - LINEAR and TIME-INVARIANT
- ALL LTI systems have h[n] & use convolution

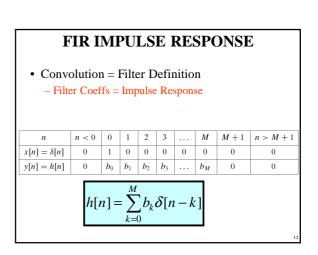




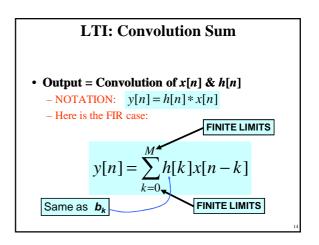


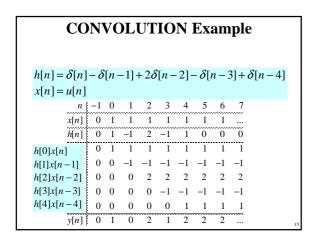


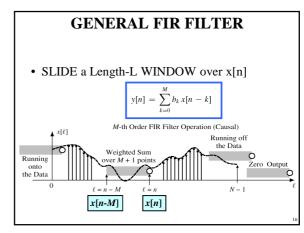


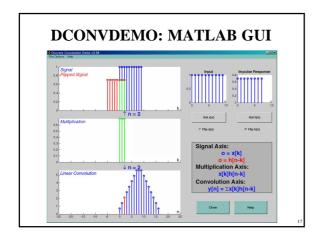


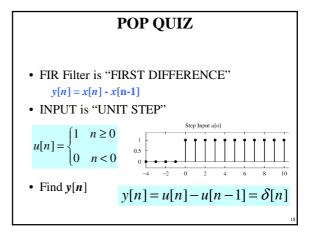
• Use SHIFTED IMPULSES to write h[n] $h[n] = \delta[n] - \delta[n-1] + 2\delta[n-2] - \delta[n-3] + \delta[n-4]$ $b_k = \{1, -1, 2, -1, 1\}$



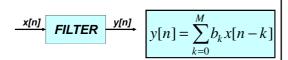




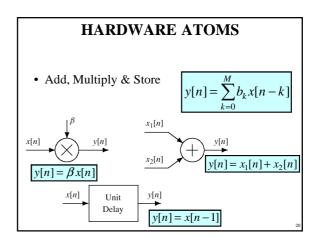


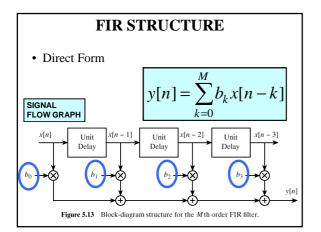


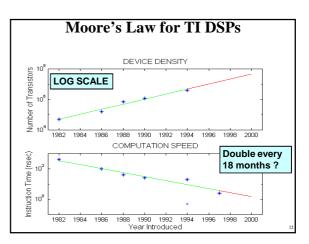
HARDWARE STRUCTURES



- INTERNAL STRUCTURE of "FILTER"
 - WHAT COMPONENTS ARE NEEDED?
 - HOW DO WE "HOOK" THEM TOGETHER?
- SIGNAL FLOW GRAPH NOTATION







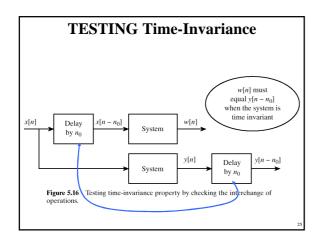
SYSTEM PROPERTIES



- MATHEMATICAL DESCRIPTION
- TIME-INVARIANCE
- LINEARITY
- CAUSALITY
 - "No output prior to input"

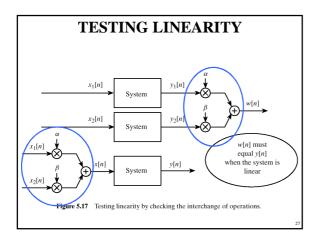
TIME-INVARIANCE

- IDEA:
 - "Time-Shifting the input will cause the same time-shift in the output"
- EQUIVALENTLY,
 - We can prove that
 - The time origin (n=0) is picked arbitrary



LINEAR SYSTEM

- LINEARITY = Two Properties
- SCALING
 - "Doubling x[n] will double y[n]"
- SUPERPOSITION:
 - "Adding two inputs gives an output that is the sum of the individual outputs"



LTI SYSTEMS

- LTI: Linear & Time-Invariant
- COMPLETELY CHARACTERIZED by:
 - IMPULSE RESPONSE h[n]
 - **CONVOLUTION**: y[n] = x[n]*h[n]
 - The "rule" defining the system can ALWAYS be rewritten as convolution
- FIR Example: h[n] is same as b_k

POP QUIZ

- FIR Filter is "FIRST DIFFERENCE"
 - -y[n] = x[n] x[n-1]
- Write output as a convolution
 - Need impulse response

$$h[n] = \delta[n] - \delta[n-1]$$

- Then, another way to compute the output:

$$y[n] = (\delta[n] - \delta[n-1]) * x[n]$$

• Does the order of S_1 & S_2 matter? • NO, LTI SYSTEMS can be rearranged !!! • WHAT ARE THE FILTER COEFFS? $\{b_k\}$ x[n] s_1 s_2 s_3 s_4 Figure 5.19 A Cascade of Two LTI Systems.

