EE327 Digital Signal Processing Convolution Yasser F. O. Mohammad

REMINDER 1:Linear Systems

- Linear = Homogeneous+Additive
- Homogeneity

• If $X[n] \rightarrow Y[n]$ then $k X[n] \rightarrow k Y[n]$

- Additive
 - If $X_1[n] \rightarrow Y_1[n] \text{ and } X_2[n] \rightarrow Y_2[n]$ then $X_1[n] + X_2[n] \rightarrow Y_1[n] + Y_2[n]$

• Most DSP linear systems are also *shift invariant* (LTI)



System

y[n]

x[n]

REMINDER 2: Sinusoidal Fidelity

- Linear system \rightarrow sinusoidal output for sinusoidal input
- Sinusoidal Fidelity → Linear System
 - (e.g. phase Lock Loop)
- This is why we can work with AC circuits using only two numbers (amplitude and phase)
- This is why Fourier Analysis is important
- This is partially why Linear Systems are important
- This is why you cannot see DSP without *sin*

REMINDER 3: Fundamental Concept of DSP



REMINDER 4: Impulse and Step Decompositions



What is convolution?

- A mathematical operation that takes two signals and produces a third one.
 - X[n]*Y[n]=Z[n]
- For us:
 - A way to get the output signal given the input signal and a representation of system function

From now one we will deal only with discrete signals if not otherwise specified

Delta function

• Delta function=Unit impulse = $\delta[n]$

$$\delta[n] = \begin{cases} 1 & n = 0 \\ 0 & otherwise \end{cases}$$

Impulse Response

Describes a SYSTEM not a signal

- We use *h*[*n*] for it
- Gives the output signal if the input to the system was a unit impulse



Other names of impulse response

- Filters
 - Filter Kernel
 - Kernel
 - Convolution Kernel
- Image processing
 - Point Spread Function

Why impulse response is important?

- It COMPLETELY describes systems FUNCTION
 - Any input can be decomposed into an impulse train
 - Linearity → Superposition → Any input
 - [Usually] Shift invariance → Any time



How to calculate the output



- Input length = N
- Impulse Response length = M
- Output length = N+M-1
- For example a 81 points input convolved with a 31 points impulse response gives 111 points output

Examples



b. High-pass Filter



More Examples

a. Inverting Attenuator



b. Discrete Derivative







Two ways to understand it

- Input Signal Viewpoint (Input Side Algorithm)
 - How each input impulse contributes to the output signal.
 - Good for your understanding
- Output Signal Viewpoint (Output Side Algorithm)
 - How each output impulse is calculated from input signal.
 - Good for your calculator

Input Side Algorithm

- Each sample is considered a scaled impulse
- Each scaled impulse results in a scaled impulse response
- Add all scaled impulse responses together



Example Input Side Algorithm







The nine responses=Total Response



X[n]*h[n]=h[n]*X[n]







3



Input Side Algorithm

```
100 'CONVOLUTION USING THE INPUT SIDE ALGORITHM
110
120 DIM X[80]
                             'The input signal, 81 points
130 DIM H[30]
                             'The impulse response, 31 points
                             'The output signal, 111 points
140 DIM Y[110]
150
160 GOSUB XXXX
                             'Mythical subroutine to load X[] and H[]
170
180 \text{ FOR I}\% = 0 \text{ TO } 110
                             'Zero the output array
190 Y(I\%) = 0
200 NEXT I%
210
220 FOR I% = 0 TO 80
                             'Loop for each point in X[]
                             'Loop for each point in H[ ]
230 FOR J\% = 0 TO 30
240 Y[I\%+J\%] = Y[I\%+J\%] + X[I\%] *H[J\%]
250 NEXT J%
260 NEXT I%
                             '(remember, \# is multiplication in programs!)
270
                             'Mythical subroutine to store Y[]
280 GOSUB XXXX
290
300 END
```





ntribution

8 9

om x[3] h[n-3]





tribution

m x[5] h[n-5]

This is true for ANY point

x[5]×h[1],

 $x[6] \times h[o]$

h[n]

Output sample *j* is calculated As: $y[j] = \sum_{i=0}^{M-1} x[j-i]h[i]$

General Output Side Flowchart

- Flip the second signal (h[n])
- Move it over the first signal (x[n])
- Each time calculate:

$$y[j] = \sum_{i=0}^{M-1} x[j-i]h[i]$$

 Continue until first signal is finished



Example Output Side Algorithm



Boundary Effect

- At the first and last M-1 points the impulse response is not fully immersed into the signal
- These points are unreliable



Output Side Algorithm

```
100 'CONVOLUTION USING THE OUTPUT SIDE ALGORITHM
110
120 DIM X[80]
                             'The input signal, 81 points
                             'The impulse response, 31 points
130 DIM H[30]
                             'The output signal, 111 points
140 DIM Y[110]
150
                             'Mythical subroutine to load X[] and H[]
160 GOSUB XXXX
170
180 \text{ FOR I}\% = 0 \text{ TO } 110
                             'Loop for each point in Y[]
190 Y[I\%] = 0
                             'Zero the sample in the output array
200 FOR J\% = 0 TO 30
                             'Loop for each point in H[]
210 IF (I\%-J\% < 0)
                             THEN GOTO 240
220 IF (I\%-J\% > 80)
                             THEN GOTO 240
230 Y(I\%) = Y(I\%) + H(J\%) * X(I\%-J\%)
240 NEXT J%
250 NEXT I%
260
270 GOSUB XXXX
                             'Mythical subroutine to store Y[]
280
290 END
```