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UNIT 2 – Part III: Discrete Random Processes

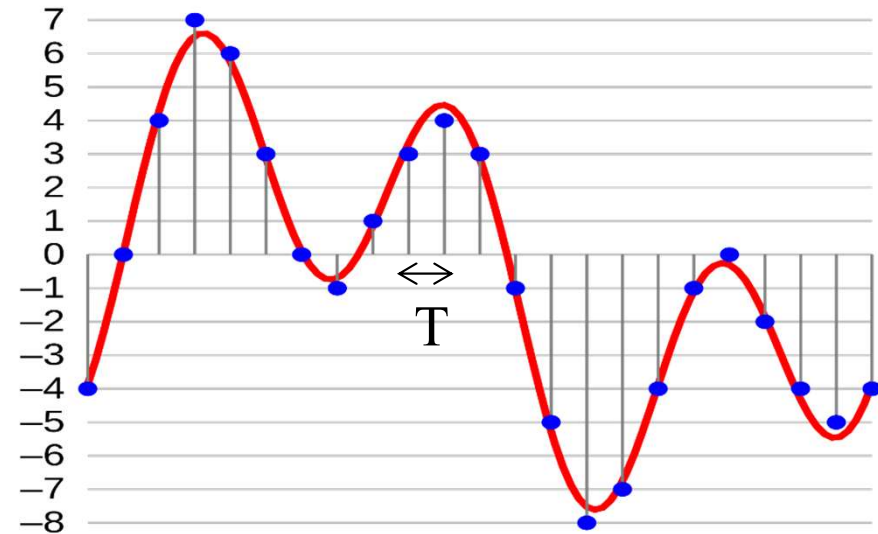
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Discrete signals

- A discrete signal $x[n]$ results from the sampling of a continuous signal $x_c(t)$ with a sampling period T ($f_s = \frac{1}{T}$)

$$x[n] = x_c(nT)$$

where n is an integer number



- The sampling frequency must be higher than twice bandwidth of the continuous signal (**Nyquist theorem**)
- $x[n]$ can now be stored in a digital system (after quantizing its amplitude as a binary number) and it can be processed by means of **digital signal processing** techniques

Fourier transform

- We can define now the equivalent to the **Laplace transform** for discrete signals: the **Z transform**

$$X(Z) = \sum_{n=-\infty}^{\infty} x[n]Z^{-n}$$

- The **Fourier transform** of discrete signals is:

$$X(e^{j\omega}) = \sum_{n=-\infty}^{\infty} x[n]e^{-j\omega n}$$

- Note that the FT is periodic with period 2π
- The Interval from 0 to 2π is equivalent to frequencies from $-\frac{f_s}{2}$ ($-\pi$ rad/s) to $\frac{f_s}{2}$ ($+\pi$ rad/s)
- The ZT and the FT are instrumental in the analysis of LTI systems
- In discrete systems, the delta function (Kronecker's delta) is defined as

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

$\delta[0]$ is not infinite!

Digital filters

- A digital filter is characterized through a discrete impulse response $h[n]$
- The output $y[n]$ of a filter with impulse response $h[n]$ and input $x[n]$ is

$$y[n] = x[n] * h[n] = \sum_{k=-\infty}^{\infty} x[k]h[n - k]$$

- It is possible to rewrite $y[n]$ and relate it to the frequency response

$$y[n] = \sum_{k=1}^N a_k y[n - k] + \sum_{k=0}^M b_k x[n - k]$$

$$H(Z) = \frac{\sum_{k=0}^M b_k Z^{-k}}{1 - \sum_{k=1}^N a_k Z^{-k}}$$

Digital filters

- Finite impulse response (FIR) filter are those whose output only depends on the input (i.e. $a_k=0$)
- The impulse response is finite in time

$$y[n] = \sum_{k=0}^M b_k x[n - k]$$

$$H(Z) = \sum_{k=0}^M b_k Z^{-k}$$

$$h[n] = \sum_{k=0}^M b_k \delta[n - k]$$

Digital filters

- For infinite impulse response (IIR) filters, the output is a function of the input and past outputs (i.e. $a_k \neq 0$)
- The impulse response is infinite in time, since it depends on the last N outputs

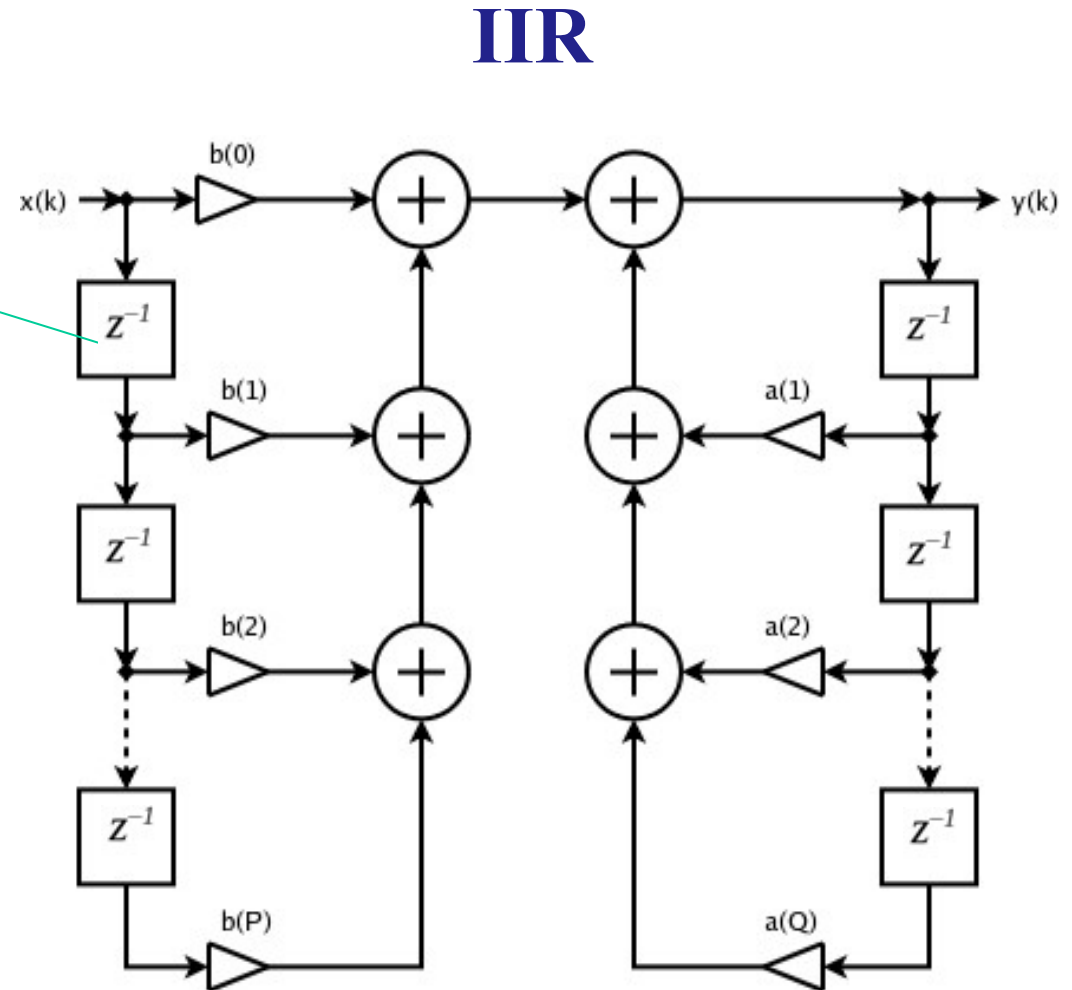
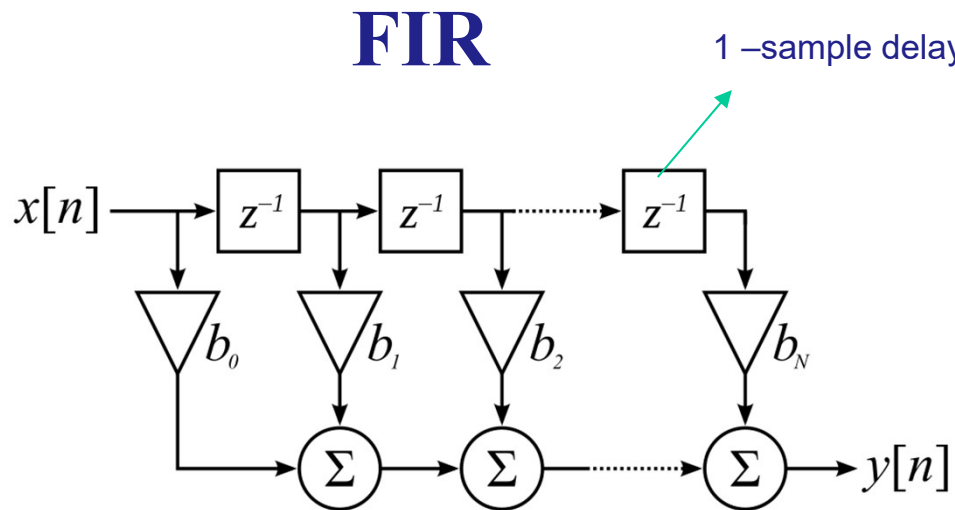
$$y[n] = \sum_{k=1}^N a_k y[n-k] + \sum_{k=0}^M b_k x[n-k]$$

$$H(Z) = \frac{\sum_{k=0}^M b_k Z^{-k}}{1 - \sum_{k=1}^N a_k Z^{-k}}$$

$$h[n] = \sum_{k=1}^N a_k y[n-k] + \sum_{k=0}^M b_k \delta[n-k]$$

Digital filters

- Example of filter structures



Discrete random processes

- Given two w.s.s. discrete random process $X[n]$ and $Y[n] = X[n] * h[n]$

$$E[X[n]] = \bar{X}$$

$$E[Y[n]] = \sum_{k=-\infty}^{\infty} h[k]E[X[n-k]] =$$

$$\bar{X} \sum_{k=-\infty}^{\infty} h[k] = \bar{X}H(e^{j0})$$

Discrete random processes

$$R_{YY}[\tau] = E[Y[n]Y[n + \tau]] = \sum_{k=-\infty}^{\infty} h[k] \sum_{r=-\infty}^{\infty} h[r] R_{XX}[\tau + k - r]$$

$$S_{YY}(\omega) = |H(e^{j\omega})|^2 S_{XX}(\omega)$$

Note the new
integral limits

$$\begin{aligned} P_{YY} = R_{YY}[0] &= \frac{1}{2\pi} \int_{-\pi}^{\pi} S_{YY}(\omega) d\omega \\ &= \frac{1}{2\pi} \int_{-\pi}^{\pi} |H(e^{j\omega})|^2 S_{XX}(\omega) d\omega \end{aligned}$$

Discrete random processes

- Given If $X[n]$ is a white noise, then

$$R_{XX}[\tau] = \frac{N_0}{2} \delta[n]$$

$$S_{XX}(\omega) = \frac{N_0}{2}$$

$$P_{XX} = R_{XX}[0] = \frac{N_0}{2}$$



Note that the power of a discrete White noise is **not** infinite

$$R_{YY}[\tau] = \frac{N_0}{2} \sum_{k=-\infty}^{\infty} h[k]h[k + \tau]$$

$$S_{YY}(\omega) = |H(e^{j\omega})|^2 S_{XX}(\omega) = |H(e^{j\omega})|^2 \frac{N_0}{2}$$

$$P_{YY} = R_{YY}[0] = R_{XX}[0] \sum_{k=-\infty}^{\infty} h[k]^2 =$$

$$P_{XX} \sum_{k=-\infty}^{\infty} h[k]^2 =$$

$$\frac{N_0}{2} \sum_{k=-\infty}^{\infty} h[k]^2 =$$

$$\left(\frac{N_0}{2}\right) \left(\frac{1}{2\pi} \int_{-\pi}^{\pi} |H(e^{j\omega})|^2 d\omega\right)$$



Parseval's theorem

Discrete random processes

- The time averages can be computed as

$$A[x[n]] = \lim_{L \rightarrow \infty} \frac{1}{2L+1} \sum_{k=-L}^L x[k]$$

$$\mathbb{R}_{xx}[x[n]x[n + \tau]] = \lim_{L \rightarrow \infty} \frac{1}{2L+1} \sum_{k=-L}^L x[k] x[k + \tau]$$

SUMMARY

- Discrete signals
- Digital filters
- Discrete random processes
 - Mean, autocorrelation, power spectrum
 - LTI systems
 - Time averages