# ECE4634 Digital Communications Fall 2007

Instructor: R. Michael Buehrer

Lecture #9: Digital Pulse

Modulation: Line Codes







- The objectives of this lecture are
  - to introduce ways of mapping data bits (1's and 0's) to waveforms which is termed a line code
  - to derive the Power Spectral Density of common line codes
  - to present properties of common line codes

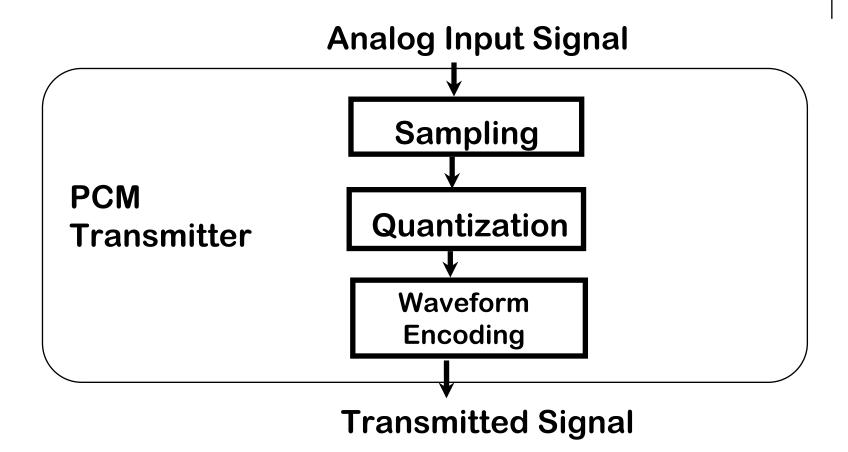
#### **Overview**



- Digital communication systems transmit a finite number of messages (typically a train of pulses) from a transmitter to the receiver.
- These waveforms represent the bit stream.
- The bits can either represent an analog signal (after analog-todigital conversion) or can be binary data.
- The data bits are mapped to waveforms (i.e., the messages) for transmission.
- Previously we assumed that bits were mapped to rectangular pulses
- Today we will study several different potential waveforms (sometimes termed "line codes") and their properties.
- What to read Section 5.9

### Structure of Digital Communications Transmitter





### Digital Signaling and Line Codes



- We learned that PCM converts an analog signal into a serial stream of 1's and 0's to encode pulses (or waveforms) for transmission.
- We have not yet discussed the properties of these pulses.
- The mapping of the bits to waveforms or pulses is the essence of digital signaling.
- Today we look at several different mappings from {0,1} to signaling formats called *line codes*.





- The simplest method of digital signaling is binary signaling
- For binary data,  $b_k = \{0,1\}$

Note  $b_k = \{-1, +1\}$  is also possible

$$w(t) = \sum_{k=-\infty}^{\infty} b_k p(t - kT)$$

- where p(t) is a pulse and T is the pulse duration.
- One simple pulse is a square pulse:

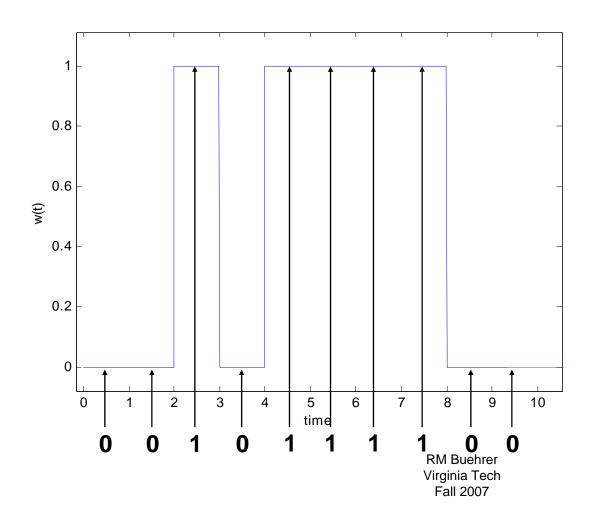
$$p(t) = \begin{cases} \frac{1}{\sqrt{T}} & T/2 < t < T/2 \\ 0 & \text{else} \end{cases}$$
RM Buehrer Virginia Tech

Fall 2007





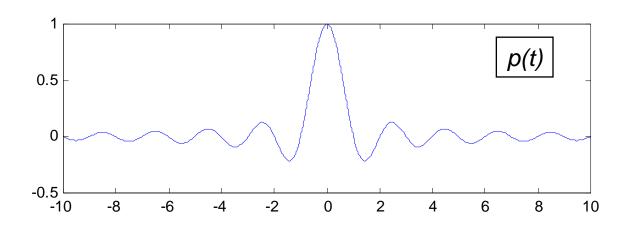
Data = [0,0,1,0,1,1,1,1,0,0]



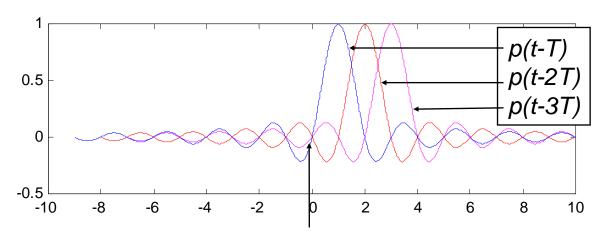
- Data controls pulse amplitude.
- Pulses are nonoverlapping
- Receiver
   determines data
   by examining a
   sample within
   the pulse width
   (note that with
   square pulse the
   exact sampling
   time is irrelevant

#### **Example: Sinc Pulses**





- Sinc pulses are a second possibility.
- They require less bandwidth than square pulses
- Unfortunately they are non-causal



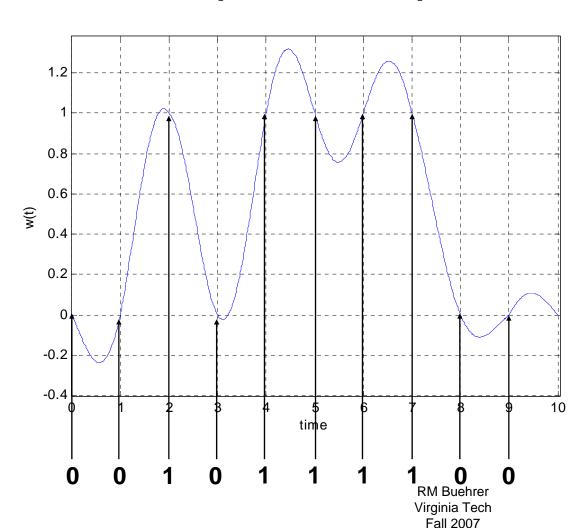
$$p(t) = \frac{\sin\left(\frac{\pi}{T}t\right)}{\frac{\pi}{T}t}$$

Note: Pulses go to zero every T seconds





Data = [0,0,1,0,1,1,1,1,0,0]



- Data controls pulse amplitude.
- Pulses overlap, but go to zero every symbol period
- Receiver determines data by examining the sample at specific sampling time (note that with sinc pulses the exact sampling time is important to avoid interference between pulses)





- With binary signaling each pulse is modulated by a single bit. Thus, the symbol rate is equal to the bit rate.
- One method of reducing bandwidth is to map more than one bit to each pulse.
- This can be accomplished by allowing the amplitude to take on L > 2 values where L is a power of 2
  - Groups of n bits are mapped into one of L=2<sup>n</sup> levels.
  - Ex: 4-ary signaling, 8-ary signaling



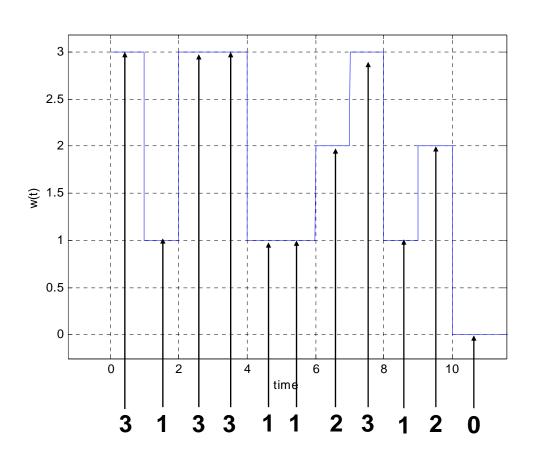
#### Multi-level Signaling (cont.)

$$w(t) = \sum_{k=-\infty}^{\infty} w_k p(t - kT)$$

- Ex: 4-ary signaling  $\rightarrow$   $w_k = \{0,1,2,3\}$ 
  - 00 → 0
  - 01 → 1
  - 10 → 2
  - 11 → 3





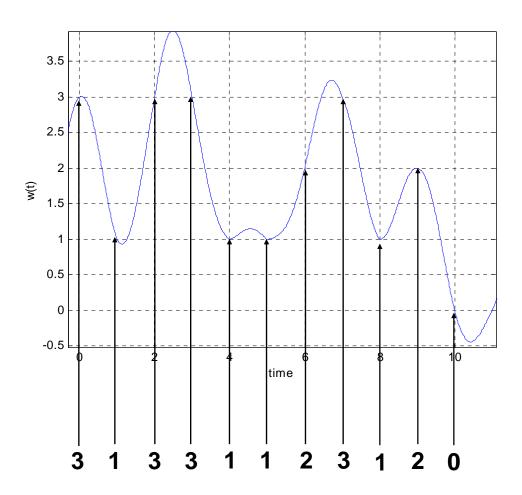


- Data controls pulse amplitude.
- Pulses are nonoverlapping
- Receiver
   determines data
   by examining a
   sample within
   the pulse width
   (note that with
   square pulse the
   exact sampling
   time is irrelevant





- Data controls pulse amplitude.
- Pulses overlap, but go to zero every symbol period
- Receiver
   determines data
   by examining the
   sample at specific
   sampling time
   (note that with sinc
   pulse the exact
   sampling time is
   important to avoid
   interference
   between pulses)



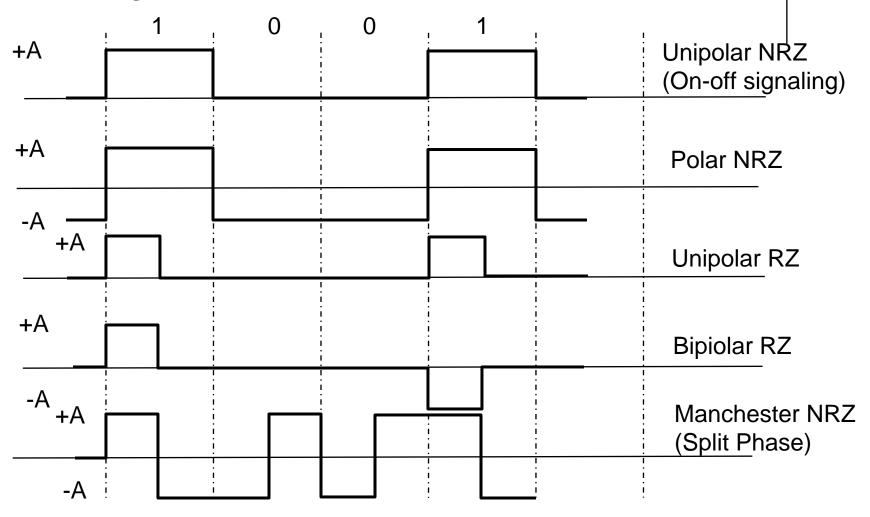


#### **Binary Line Codes**

- When binary signaling is used, there are various ways to create waveforms with a square pulse shape (so far we have assumed unipolar NRZ signaling).
- These various signaling formats are called line codes.
- There are two major categories of line codes
  - Return to zero codes (RZ)
    - The signal returns to zero for ½ (or other fraction) of the pulse
  - Non-return to zero codes (NRZ)
- Codes have different
  - Spectral properties
  - Synchronization capabilities
  - Error performance

#### **Binary Line Codes**





# Power Spectral Densities (Deterministic)



Deterministic and Periodic Power Signals

$$x(t) = \sum_{n=-\infty}^{\infty} c_n e^{j2\pi n f_o t}$$

- where  $c_n$  are the Fourier Series Coefficients
- Deterministic power signals have Power Spectral Density (PSD):

$$S_{x}(f) = \sum_{n=-\infty}^{\infty} |c_{n}|^{2} \delta(f - nf_{0})$$

# Power Spectral Densities (Random)

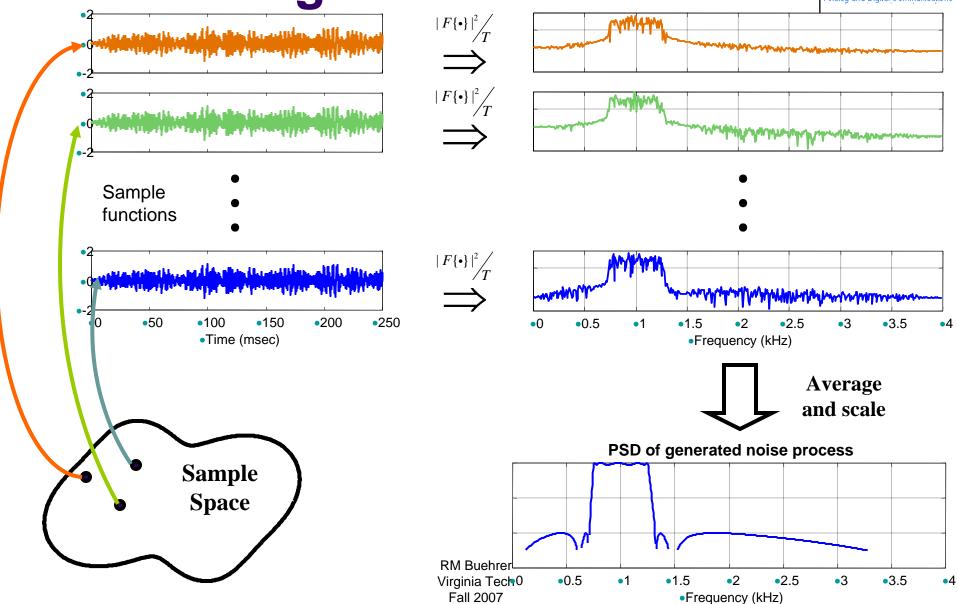


• It is easiest to talk about random signals that extend for all time (i.e., power signals). Hence almost every sample function (or realization of a stochastic process) is a power signal with a power spectral density. In fact any *stationary* random process will have infinite energy. The PSD of the random process is just the ensemble average of all sample function power spectral densities.

$$S_X(f) = E \left[ \lim_{T \to \infty} \frac{|X_T(f)|^2}{T} \right] = \lim_{T \to \infty} \frac{E[|X_T(f)|^2]}{T}$$







#### **Power Spectral Density**



- $S_x(f)$  (PSD) tells us how much power is at each frequency
- For deterministic signals, the function description x(t) tells us how the value changes with time, thus its Fourier transform X(f) gives us the spectral properties. However, random processes have a random function description
- Wiener-Khinchine Theorem
  - For a WSS process

$$S_{x}(f) = F \left\{ R_{x}(\tau) \right\}$$
$$= \int_{-\infty}^{\infty} R_{x}(\tau) e^{-j2\pi f \tau} d\tau$$

- The autocorrelation function  $R(\tau) = \int_{-\infty}^{\infty} w(t)w(t+\tau)dt$  tells us how the value is *expected* to change with time rather than the exact change.
- Power spectral density and autocorrelation are a Fourier Transform pair

#### Power Spectral Density for Digital Modulated Pulse Train



Consider a digitally modulated pulse train:

$$x(t) = \sum_{n=0}^{\infty} a_n f\left(t - nT_s\right)$$

where f(t) is the pulse shape and  $a_n$  are the digital values.

We can show (see Appendix) that the PSD of this signal is:

$$S_{x}(f) = \frac{\sigma_{a}^{2}}{T_{s}} |F(f)|^{2} + \frac{m_{a}^{2}}{T_{s}^{2}} \sum_{n=-\infty}^{\infty} \left| F\left(\frac{n}{T_{s}}\right) \right|^{2} \delta\left(f - \frac{n}{T_{s}}\right)$$
continuous

- PSD has continuous portion which is dependent on the pulse shape f(t) and a discrete portion which is also dependent on the data mean value and variance.
- This will become very important when we examine the bandwidth requirements of digitally modulated signals





+A 0 0 1

**Unipolar NRZ** 

$$m_a = \frac{A}{2}$$

$$\sigma_a^2 = \frac{A^2}{4}$$

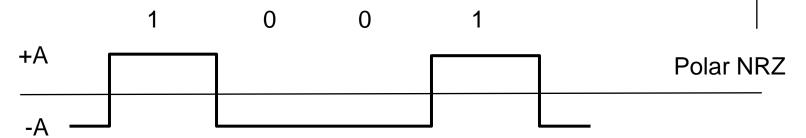
$$F(f) = T_b \frac{\sin(\pi f T_b)}{\pi f T_b}$$

$$P(f) = \frac{A^{2}T_{b}}{4} \left( \frac{\sin(\pi f T_{b})}{\pi f T_{b}} \right)^{2} \left[ 1 + \underbrace{\frac{1}{T_{b}} \delta(f)}_{\text{due to pulse shape}} \right]$$

RM Buehrer Virginia Tech Fall 2007







$$m_a = 0$$
$$\sigma_a^2 = A^2$$

$$F(f) = T_b \frac{\sin(\pi f T_b)}{\pi f T_b}$$

Same as Polar RZ except that DC component is eliminated.

$$P(f) = A^{2}T_{b} \left( \frac{\sin(\pi f T_{b})}{\pi f T_{b}} \right)^{2}$$

Error performance is superior as compared to Polar RZ signaling. This is because no power is wasted in DC component.

#### **Unipolar RZ**



0 0

+A \_\_\_\_\_\_

Uniolar RZ

$$m_a = \frac{A}{2}$$

$$\sigma_a^2 = \frac{A^2}{4}$$

$$F(f) = \frac{T_b}{2} \frac{\sin(\pi f T_b / 2)}{\pi f T_b / 2}$$

Note: Discrete terms occur at multiples of  $1/T_b$ . However, the pulse shape forces the spectrum to zero at multiples of  $2/T_b$ .

$$P(f) = \frac{A^2 T_b}{16} \left( \frac{\sin(\pi f T_b / 2)}{\pi f T_b / 2} \right)^2 \left[ 1 + \frac{1}{T_b} \sum_{n = -\infty}^{\infty} \delta \left( f - \frac{n}{T_b} \right) \right]$$
due to pulse shape
$$1 + \frac{1}{T_b} \sum_{n = -\infty}^{\infty} \delta \left( f - \frac{n}{T_b} \right)$$
discrete terms due to correlation in data





$$m_a = 0$$

$$\sigma_a^2 = \frac{A^2}{2}$$

Note: Discrete terms eliminated by removing correlation in data (including DC component). sin<sup>2</sup>(x) term forces spectrum to zero at DC.

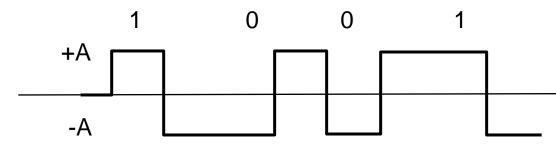
$$F(f) = \frac{T_b}{2} \frac{\sin(\pi f T_b/2)}{\pi f T_b/2}$$

$$P(f) = \frac{A^2 T_b}{4} \left( \frac{\sin(\pi f T_b / 2)}{\pi f T_b / 2} \right)^2 \sin^2(\pi f T_b)$$

RM Buehrer Virginia Tech Fall 2007







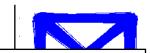
Manchester NRZ

$$m_a = 0$$
$$\sigma_a^2 = A^2$$

$$F(f) = jT_b \frac{\sin(\pi f T_b/2)}{\pi f T_b/2} \sin\left(\frac{2\pi f T_b}{4}\right)$$

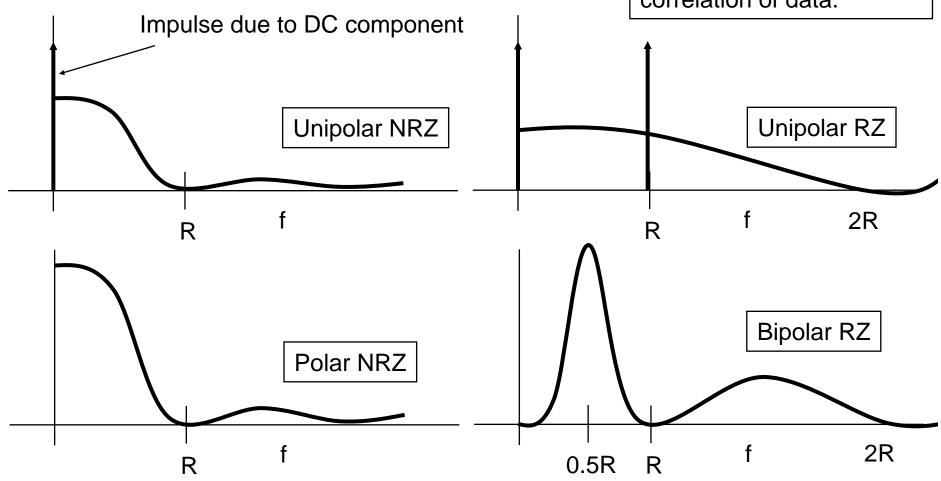
sin<sup>2</sup>(x) term forces spectrum to zero at DC. First null BW twice that of bipolar RZ

$$P(f) = A^{2}T_{b} \left( \frac{\sin(\pi f T_{b}/2)}{\pi f T_{b}/2} \right)^{2} \sin^{2}(\pi f T_{b}/2)$$



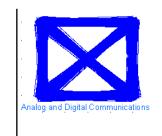
#### **Spectra of Line Codes**

Note: Spectra are dependent on pulse shape and auto-correlation of data.



26





- Self-synchronization
  - Timing information is built into the code to allow clock recovery.
  - Strings of 1's and 0's do not cause timing problems
- Spectrum
  - The spectrum must be appropriate for the channel being used.
  - Bandwidth should be as small as possible
- Error-probability
  - The code should provide low probability of bit error
  - Code should allow easy implementation of error corrections coding/decoding





- Unipolar codes
  - Have the advantage of requiring only a single positive voltage power supply.
  - Have the disadvantage of having a DC value which is less power efficient and requires channels that have DC response.
- Bipolar Codes
  - Have zero DC level provided 0's and 1's occur at the same frequency and there are not long strings of 0's or 1's.
- Manchester Codes
  - Have no DC value regardless of the number of consecutive 1's or 0's.
  - Twice the bandwidth of NRZ codes
- RZ codes
  - Have self-synchronization properties due to spectral lines (or ability to produce spectral lines) at f = R.
  - Have twice the bandwidth of NRZ signals





- PCM is a common method of converting an analog signal to a digital waveform
- Sampling and quantization converts analog waveforms into a series of bits
- There are several methods of mapping the bits to digital waveforms
  - Binary vs. multi-level signaling
  - Various pulse shapes
  - Various line codes
- Next class we will look more closely at the effect of pulse shape

#### **Appendix**

Derivation of the PSD for a digitally modulated pulse train



#### **Power Spectral Density for** Digitally Modulated Pulse Train



Let us define a general digitally modulated pulse train as  $x(t) = \sum_{n=0}^{\infty} a_n f(t - nT_s)$ 

where  $a_n$  is the sequence of data values, f(t) is the pulse shape used and  $T_s$  is the symbol duration. This is sometimes called a modulated pulse train.

Now the Power Spectral Density can be found as

$$P_{x}(f) = \lim_{T \to \infty} \left( \frac{\left| X_{T}(f) \right|^{2}}{T} \right)$$

Where 
$$X_T(f) = \int_{-T/2}^{T/2} x(t)e^{-j2\pi ft}dt$$
 and  $\overline{X(f)} = E[X(f)] = \text{ensemble average}$ 

From our signal definition we have 
$$X_T(f) = \int_{-T/2}^{T/2} \sum_{n=-\infty}^{\infty} a_n f\left(t - nT_s\right) e^{-j2\pi f t} dt$$
 
$$= \sum_{n=-N}^{N} a_n \int_{-\infty}^{\infty} f\left(t - nT_s\right) e^{-j2\pi f t} dt$$
 
$$\boxed{T = (2N+1)T_s}$$



Continuing from the previous slide

$$X_{T}(f) = \sum_{n=-N}^{N} a_{n} \int_{-\infty}^{\infty} f(t - nT_{s}) e^{-j2\pi ft} dt$$

$$= \sum_{n=-N}^{N} a_{n} F(f) e^{-j2\pi fnT_{s}}$$

$$= F(f) \sum_{n=-N}^{N} a_{n} e^{-j2\pi fnT_{s}}$$

Returning to our definition for PSD  $P_x(f) = \lim_{T \to \infty} \left( \frac{\overline{\left| X_T(f) \right|^2}}{T} \right)$ 

$$= \lim_{T \to \infty} \left( \frac{\left| F(f) \sum_{n=-N}^{N} a_n e^{-j2\pi f n T_s} \right|^2}{T} \right)$$

$$= |F(f)|^2 \lim_{N \to \infty} \left( \frac{1}{(2N+1)T_s} \sum_{n=-N}^{N} \sum_{m=-N}^{N} \overline{a_n a_m} e^{-j2\pi f(n-m)T_s} \right)$$

Fall 2007



Defining 
$$R(k) = a_n a_{n+k}$$
 and  $m = n+k$ 

$$\begin{split} P_{x}(f) &= \left| F(f) \right|^{2} \lim_{N \to \infty} \left( \frac{1}{(2N+1)T_{s}} \sum_{n=-N}^{N} \sum_{m=-N}^{N} \overline{a_{n}} a_{m} e^{-j2\pi f(n-m)T_{s}} \right) \\ &= \left| F(f) \right|^{2} \lim_{N \to \infty} \left( \frac{1}{(2N+1)T_{s}} \sum_{n=-N}^{N} \sum_{k=-N-n}^{N-n} R(k) e^{-j2\pi fkT_{s}} \right) \\ &= \frac{\left| F(f) \right|^{2}}{T_{s}} \lim_{N \to \infty} \left( \frac{(2N+1)}{(2N+1)} \sum_{k=-N}^{N} R(k) e^{-j2\pi fkT_{s}} \right) \\ &= \frac{\left| F(f) \right|^{2}}{T_{s}} \sum_{k=-\infty}^{\infty} R(k) e^{-j2\pi fkT_{s}} \\ &= \frac{\left| F(f) \right|^{2}}{T_{s}^{2}} \sum_{k=-\infty}^{\infty} R(k) \delta \left( f - \frac{k}{T_{s}} \right) \\ &= \frac{\sum_{k=-\infty}^{\infty} e^{-j2\pi fkT_{s}} = \frac{1}{T_{s}} \sum_{k=-\infty}^{\infty} \delta \left( f - \frac{k}{T_{s}} \right) \end{split}$$

RM Buehrer Virginia Tech Fall 2007



For uncorrelated\* data

$$R(k) = \begin{cases} \overline{a_n^2} & k = 0 \\ \overline{a_n a_{n+k}} & k \neq 0 \end{cases} = \begin{cases} \sigma_a^2 + m_a^2 & k = 0 \\ m_a^2 & k \neq 0 \end{cases}$$

Thus,

$$P_{x}(f) = \frac{\left|F(f)\right|^{2}}{T_{s}} \sum_{k=-\infty}^{\infty} R(k)e^{j2\pi fkT_{s}}$$

$$= \frac{\left|F(f)\right|^{2}}{T_{s}} \left(\sigma_{a}^{2} + m_{a}^{2} \sum_{n=-\infty}^{\infty} e^{j2\pi fkT_{s}}\right)$$

$$= \frac{\sigma_{a}^{2}}{T_{s}} \left|F(f)\right|^{2} + \frac{m_{a}^{2}}{T_{s}^{2}} \left|F(f)\right|^{2} \sum_{n=-\infty}^{\infty} \delta\left(f - \frac{n}{T_{s}}\right)$$

$$= \frac{\sigma_{a}^{2}}{T_{s}} \left|F(f)\right|^{2} + \frac{m_{a}^{2}}{T_{s}^{2}} \sum_{n=-\infty}^{\infty} \left|F\left(\frac{n}{T_{s}}\right)\right|^{2} \delta\left(f - \frac{n}{T_{s}}\right)$$

$$\xrightarrow{continuous} \frac{discrete}{discrete}$$

<sup>\*</sup> For uncorrelated data, E{XY} = E{X}E{Y}



$$P_{x}(f) = \frac{\sigma_{a}^{2}}{T_{s}} |F(f)|^{2} + \frac{m_{a}^{2}}{T_{s}^{2}} \sum_{n=-\infty}^{\infty} \left| F\left(\frac{n}{T_{s}}\right) \right|^{2} \delta\left(f - \frac{n}{T_{s}}\right)$$
continuous

- PSD has continuous portion which is dependent on the pulse shape f(t) and a discrete portion which is also dependent on the data mean value and variance.
- This will become very important when we examine the bandwidth requirements of digitally modulated signals