

Analogue Data / Digital Signals

- ◆ As stated before a **CODEC** is used to generate a **Digital Signal**:
 - Hence this is the device to be considered here,
 - This device was also used to encode Digital Data onto a Digital Signal:
 - Recall the use of NRZ, Bipolar AMI and Manchester techniques.
- ◆ There are two encoding methods to consider:
 - Pulse Code Modulation,
 - Delta Modulation.

Analogue Data / Digital Signals

- ◆ To encode **Analogue Data** onto a **Digital Signal** requires two steps:
 - Convert *analogue data* into *digital data*,
 - Encode the *digital data* onto a *digital signal*.
- ◆ The first step requires the use of **Sampling**:
 - This is necessary as an Analogue Data signal has an infinite amount of data. Consequently, it is not feasible to send all of the Analogue Data.
- ◆ Nyquist also did some work on sampling.

Analogue Data / Digital Signals

◆ The Sampling Theorem (from Nyquist):

“If a signal $f(t)$ is sampled at regular intervals of time and at a rate higher than twice the highest signal frequency, then the samples will contain all the information necessary for the reconstruction of the original signal.”

Pulse Code Modulation

- ◆ The original analogue signal is *sampled* at regular intervals to produce *PAMS*.
- ◆ The PAMS are *quantized* i.e. assigned a binary value.
- ◆ There are considerations to be given to the number of bits to use to represent each sample:
 - These will be explored in class.

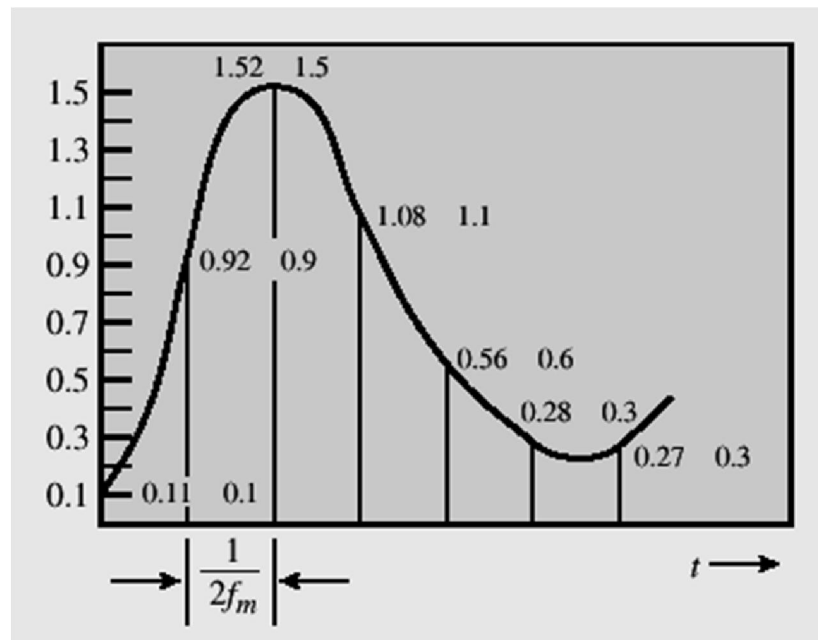
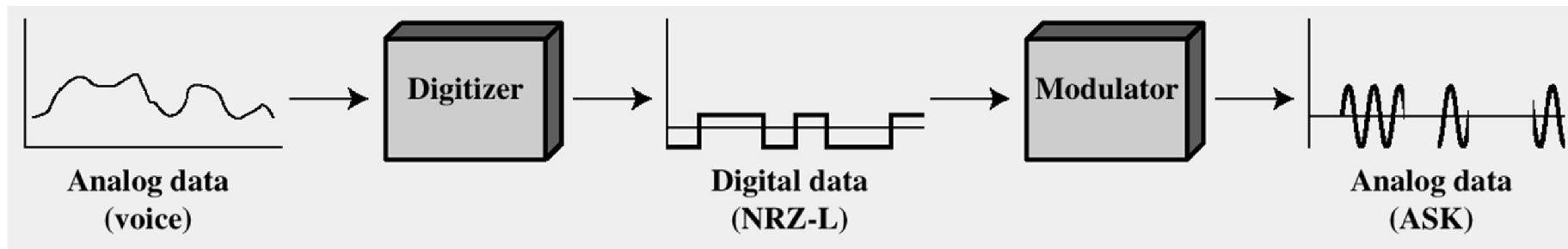
Pulse Code Modulation

- ◆ Regardless of the number of bits chosen there will always be some level of 'rounding':
 - This is known as *Quantization* which gives rise to *Quantization error* or *Quantizing noise*.
 - Hence, the original Analogue Data signal can never be truly reproduced.

Pulse Code Modulation

- ◆ Using PCM to encode voice data results in:
 - A stream of 8-bit samples,
 - Each sample is assigned an 8-bit value,
 - Resulting in a Data Rate of 64Kbps in Europe.
 - This calculation will be demonstrated in class.

Pulse Code Modulation



Digit	Binary Equivalent	PCM waveform
0	0000	Low level
1	0001	Low level, then high level
2	0010	Low level, then high level, then low level
3	0011	Low level, then high level, then high level
4	0100	Low level, then high level, then low level, then high level
5	0101	Low level, then high level, then high level, then low level
6	0110	Low level, then high level, then low level, then high level, then low level
7	0111	Low level, then high level, then high level, then high level
8	1000	High level, then low level, then low level, then low level
9	1001	High level, then low level, then low level, then high level
10	1010	High level, then low level, then high level, then low level
11	1011	High level, then low level, then high level, then high level
12	1100	High level, then high level, then low level, then low level
13	1101	High level, then high level, then low level, then high level
14	1110	High level, then high level, then high level, then low level
15	1111	High level, then high level, then high level, then high level

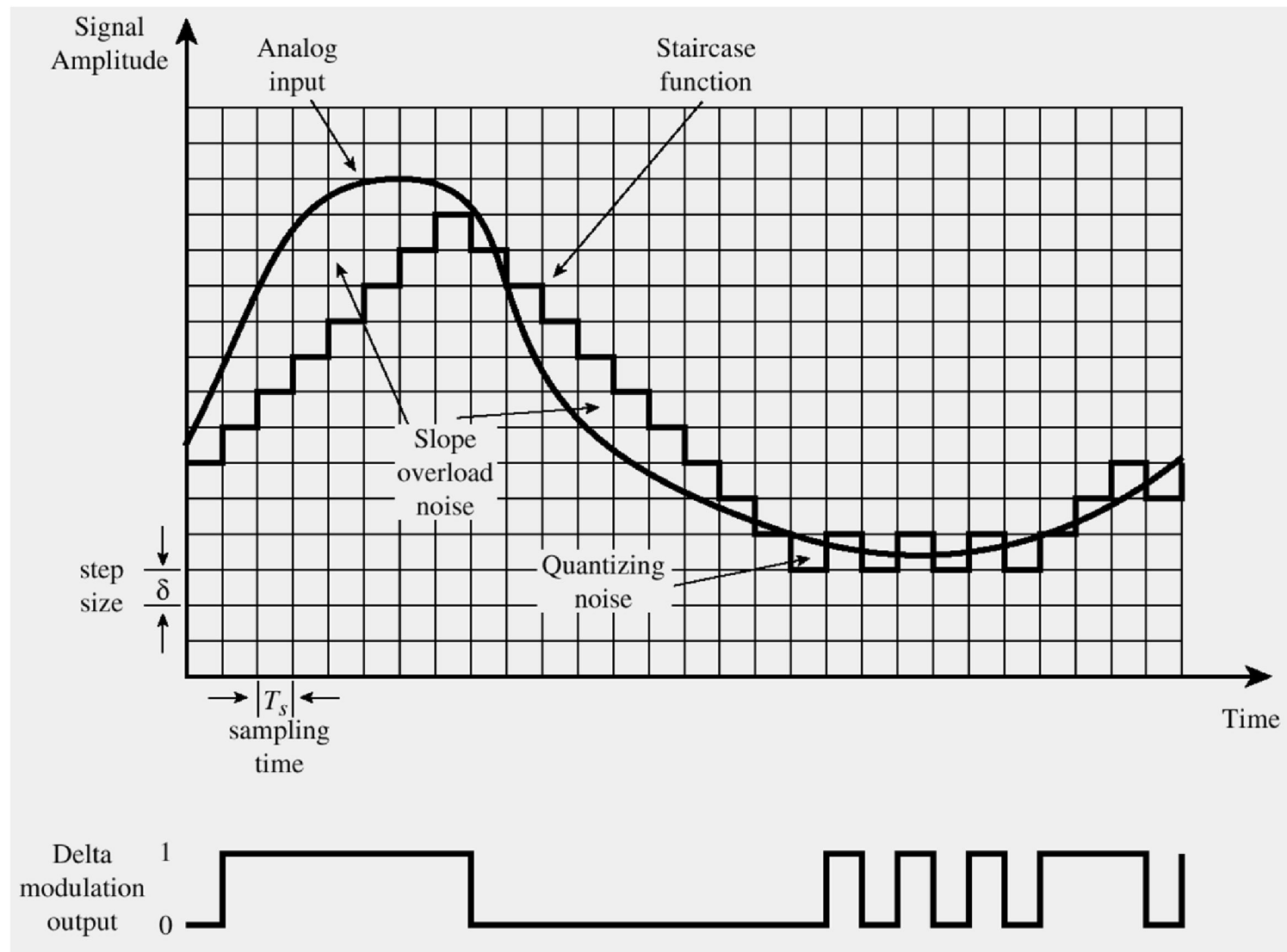
Delta Modulation

- ◆ This technique involves the use of a *staircase function*:
 - Here the analogue signal is approximated by a simple mathematical function,
 - At each sampling interval a positive or negative *quantization step* is added to the output.

Delta Modulation

- ◆ The result is a *single* binary digit for each sample indicating a positive or negative slope in the original analogue signal.
- ◆ An explanation will be given in class.

Delta Modulation



Delta Modulation

◆ Two Significant Parameters:

- Quantization step *delta* (δ)
 - large delta reduces *Slope Overload Noise*
 - small delta reduces *Quantization Noise*
- Sampling rate
 - Large rate reduces noise but increases data rate of output signal

◆ DM compared to PCM:

- DM is easier to implement
- PCM produces better SNR characteristics

Voice data: Digital V's Analogue

- ◆ Digital (PCM):
 - Good voice reproduction with 256 quantisation levels i.e. 8-bit coding
 - Required sampling rate of 8000 per second for 4000hz voice i.e. 64Kbps voice channels
 - 64Kbps voice channel requires approx. 32KHz. of BW
- ◆ Analogue (PoTS):
 - Voice channel requires approx. 4KHz. of BW
- ◆ However, digital is still preferable to analogue for the following reasons (see next slide)

Analogue Data / Digital Signals

- ◆ Popularity increasing
 - Repeaters can be used so no additive noise
 - Time-division multiplexing (TDM) can be used eliminating intermodulation noise which occurs with FDM
 - Digital signalling allows use of digital switching techniques

Analogue Data / Analogue Signals

- ◆ Analogue data can be used to modulate an analogue signal
- ◆ This is often used to:
 - Obtain a more appropriate frequency for a particular transmission
 - Allow a number of analogue signals to share a transmission medium (Frequency Division Multiplexing) – will examine later
- ◆ Three basic techniques available

Analogue Modulation Techniques

- ◆ Amplitude Modulation
 - Carrier amplitude varies in proportion to the amplitude of the analogue data
- ◆ Frequency Modulation
 - Frequency deviation is proportional to the analogue data
- ◆ Phase Modulation
 - Phase is proportional to the analogue data
- ◆ These techniques were examined previously (see Digital Data/Analogue Signal slides)