

# **Chapter 7**

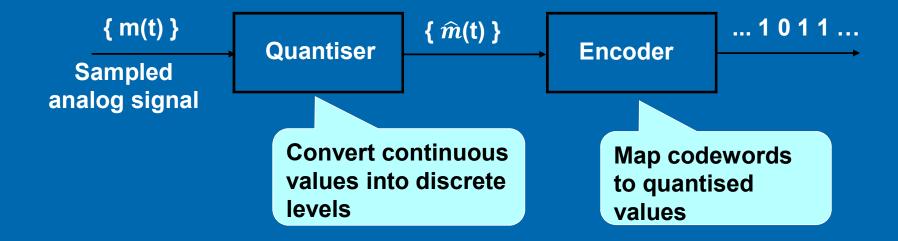
# **Analog to Digital conversion**

Part 2 of 4



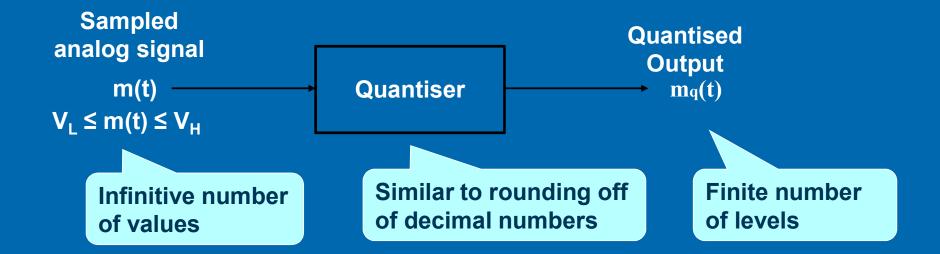


- The sampled signal from a sampling process is still an analog signal.
- To obtain a digital representation of a sampled analog signal, quantisation and encoding are required





Quantisation - a process that converts sampled analog signal into discrete levels





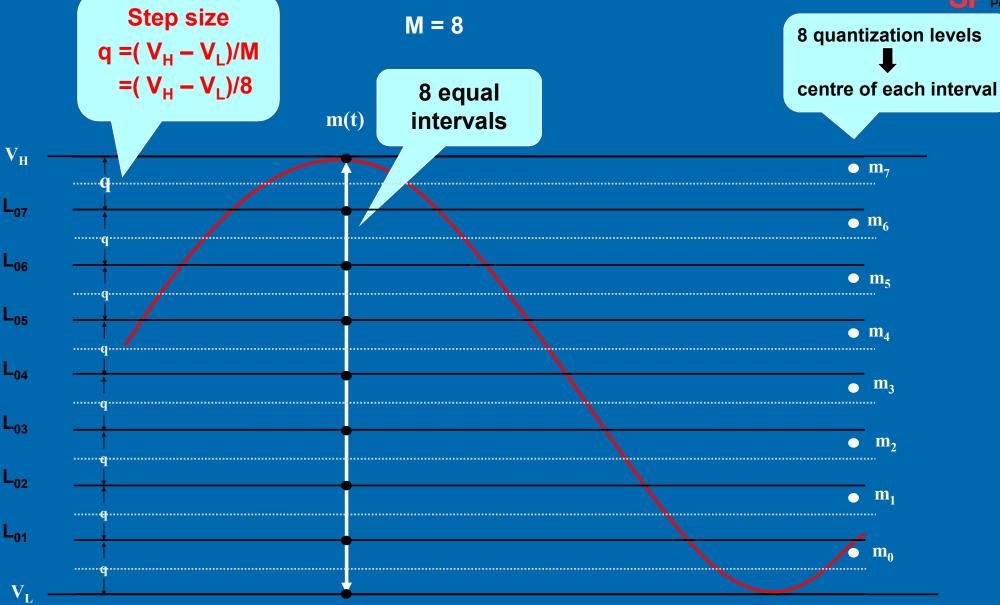
### **Quantisation process:**

1. Divide the voltage range,  $V_L$  to  $V_H$  into M equal intervals, denoted as  $L_{01}$ ,  $L_{02}$   $L_{03}$ ...  $L_{0M}$ 

The size of a intervals:  $q = (V_H - V_L) / M$  Step-size

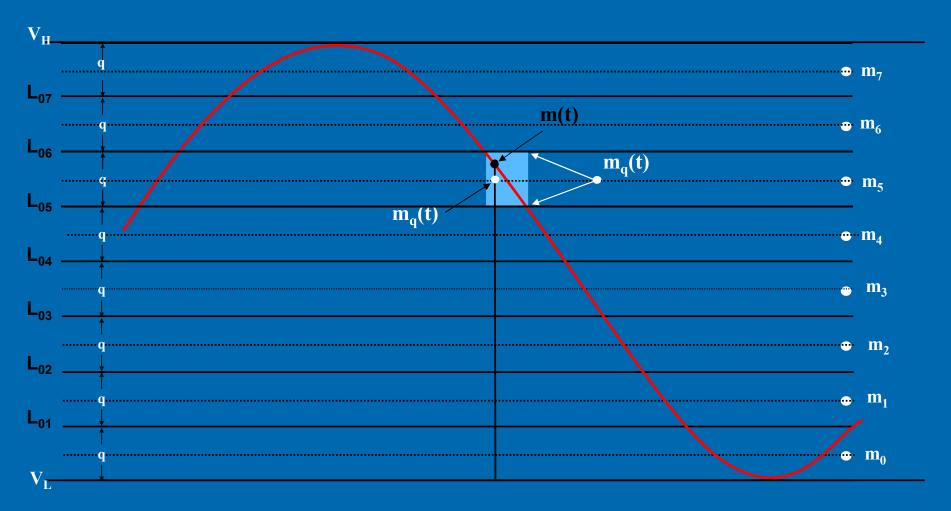
- 2. Choose the center of each interval as a quantisation level, denoted as  $m_0$ ,  $m_1$ ,  $m_2$ ...,  $m_M$ .
- 3. Represent m(t) by  $m_q(t)$ , where  $m_q(t) \in \{ m_0, m_1, m_2..., m_M \}$ 
  - At any time,  $m_q(t)$  has the value of a quantisation levels which is closest to m(t).





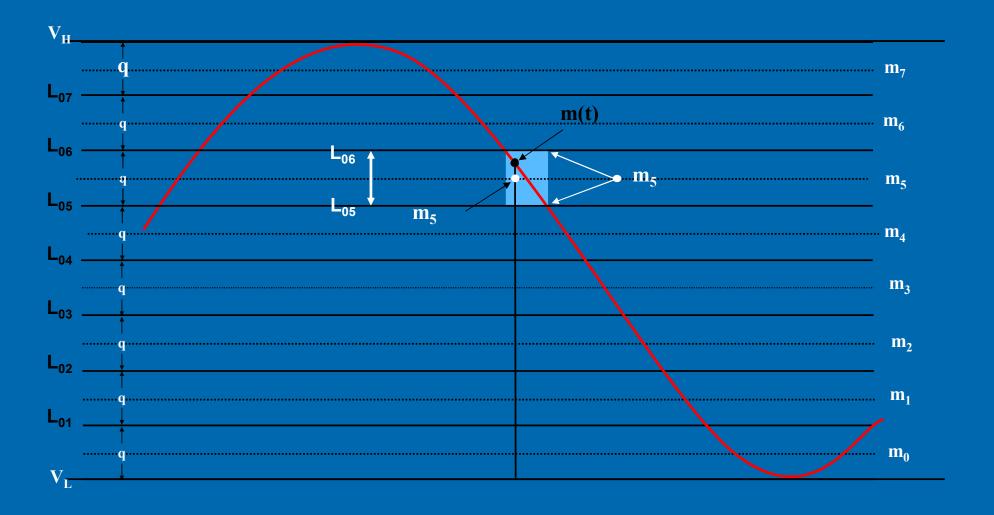


- For a sampled amplitude value m t, the quantizer rounds it up or down to make it equal to one of the 8 different quantization levels.
- The quantization level chosen is the nearest to m(t).

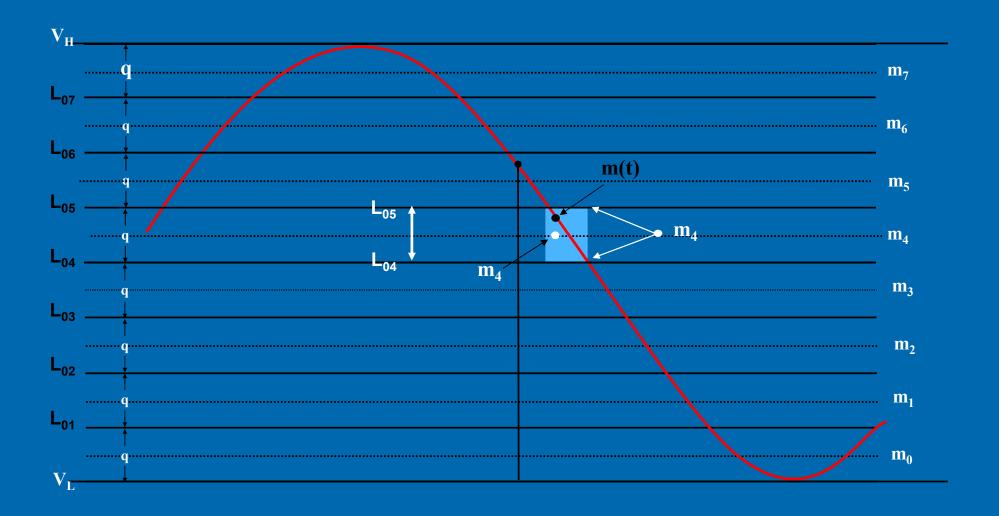




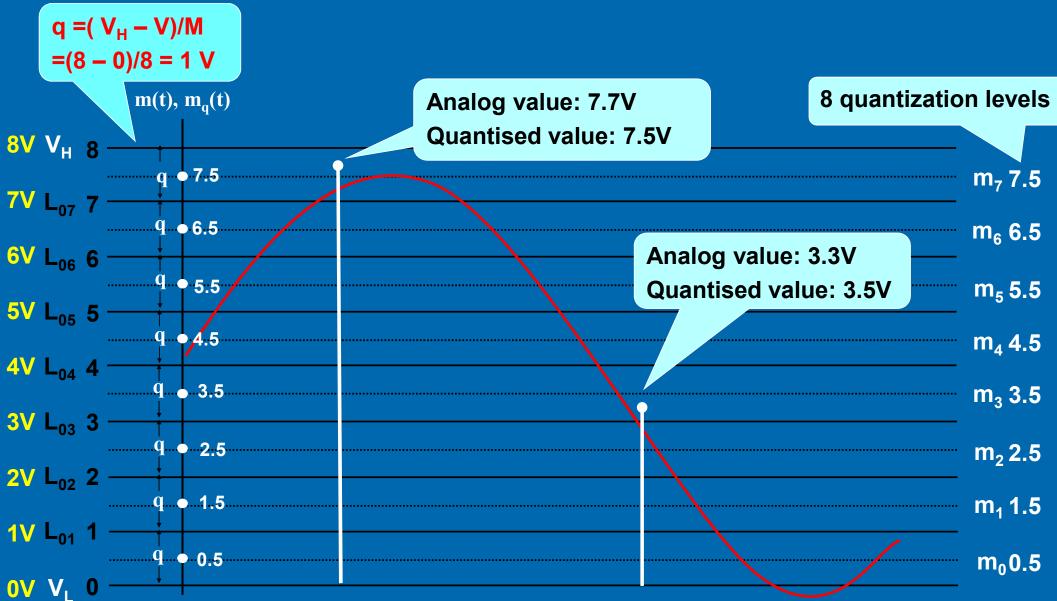




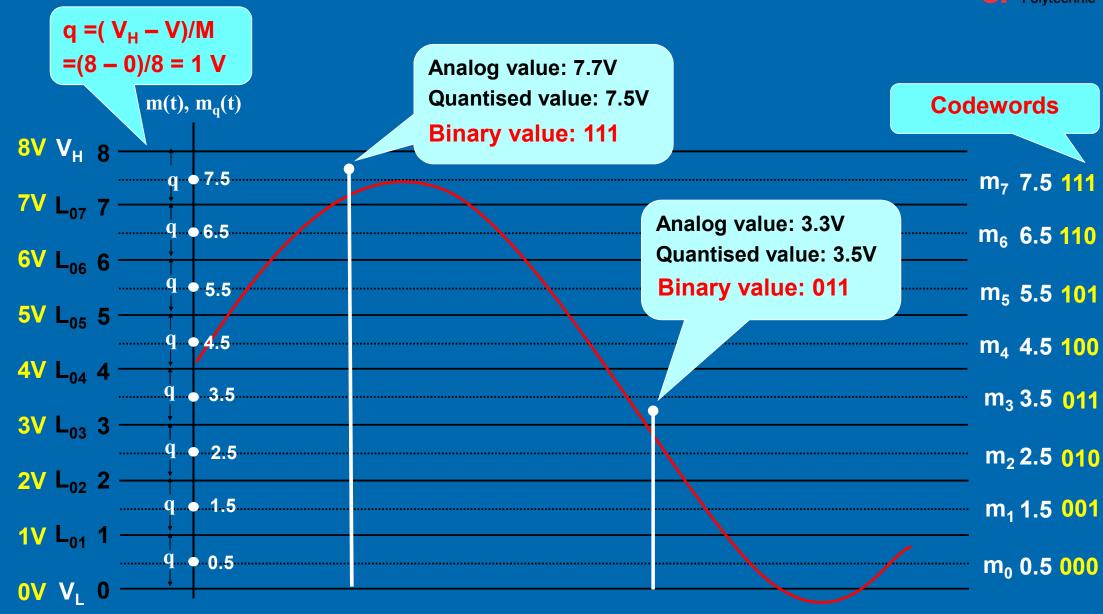




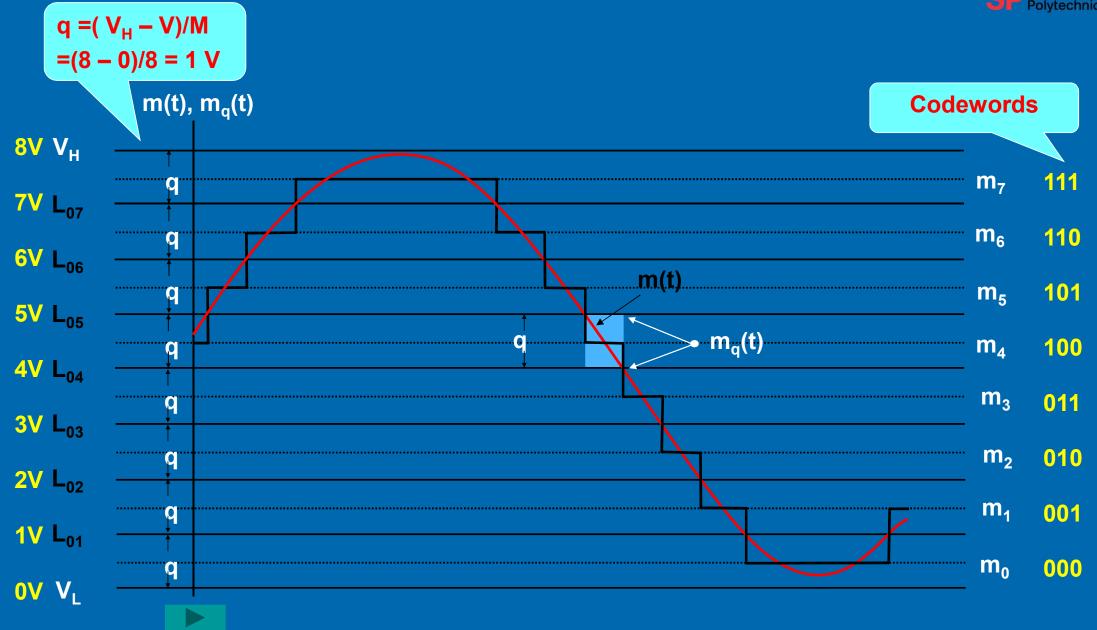
















- The quantisation process always introduces error as it approximates the sampled analog signal using quantization levels.
- The quantisation error is defined as:

Quantisation error = 
$$m_q(t)$$
 -  $m(t)$ 

• At any time instant, the quantisation error magnitude,  $|m_q(t) - m(t)|$  is equal or less than q/2.

$$|m_{q}(t) - m(t)| \le q/2$$
 or  $-q/2 \le m_{q}(t) - m(t) \le q/2$ 

The quantisation error is regarded as noise and is also called quantisation noise.

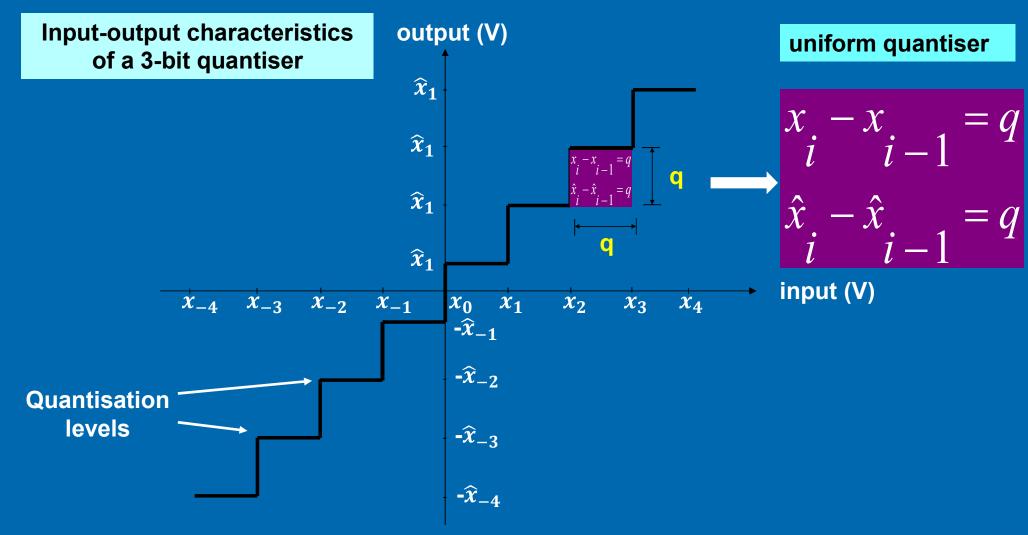


- The quality of the approximation of the quantisation process is improved by
  - reducing the step size
  - increasing the number of allowable levels.
- Depending on intended applications, different quantisation steps and levels may be chosen.
- E.g. voice telephony: 8 bits/sample 28 = 256 levels
   Audio CD: 16 bits/sample 216 = 65,536 levels
- HowStuffWorks "How DVDs Work"



**Uniform quantisation** 

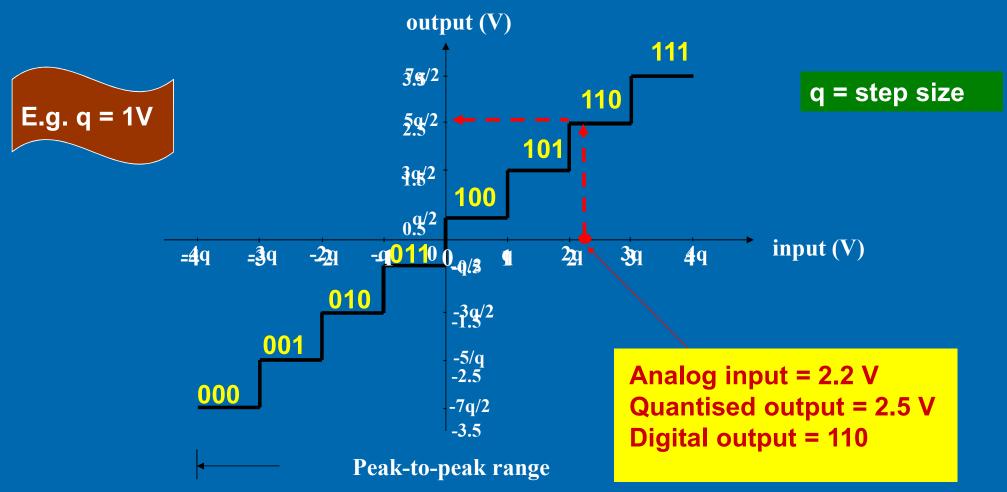
A quantiser can be defined by its input-output characteristics.





### **Uniform quantisation**

A common type of uniform quantiser characteristic: mid-riser type





### **Uniform quantisation**

- A uniform quantiser is defined by two parameters:
  - number of levels
  - step size
- The number of levels, M is generally chosen to be = 2<sup>B</sup> to make the most efficient use of B-bit binary codewords. i.e.  $M = 2^B$  E.g. B = 4,  $M = 2^4 = 16$  levels
- q and B must be chosen so as to cover the entire range of input samples which means we should set:

peak-to-peak signal amplitude = input range of quantizer, i.e.  $2 X_{max} = q2^{B}$ 

$$2 X_{max} = q2^{B}$$

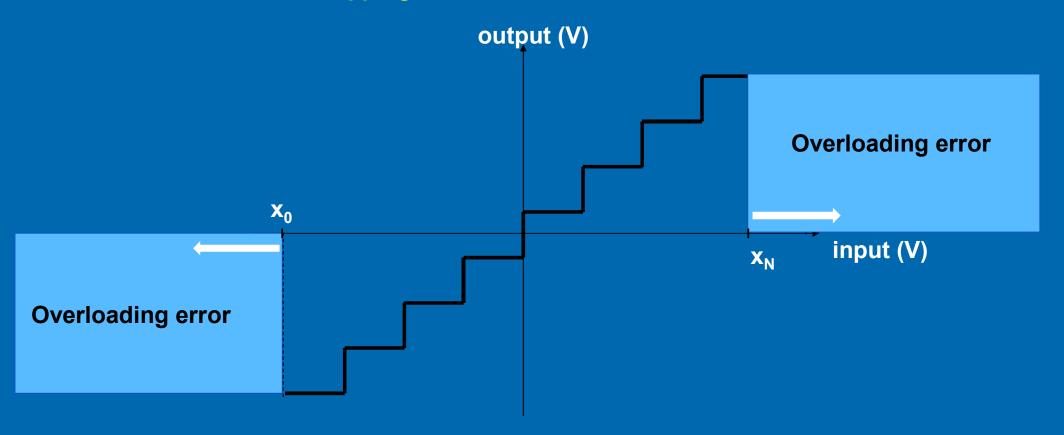
i.e. 
$$q = 2 X_{max} / 2^{B}$$

 $q = 2 X_{max} / 2^{B}$  where  $X_{max}$  is the peak signal amplitude



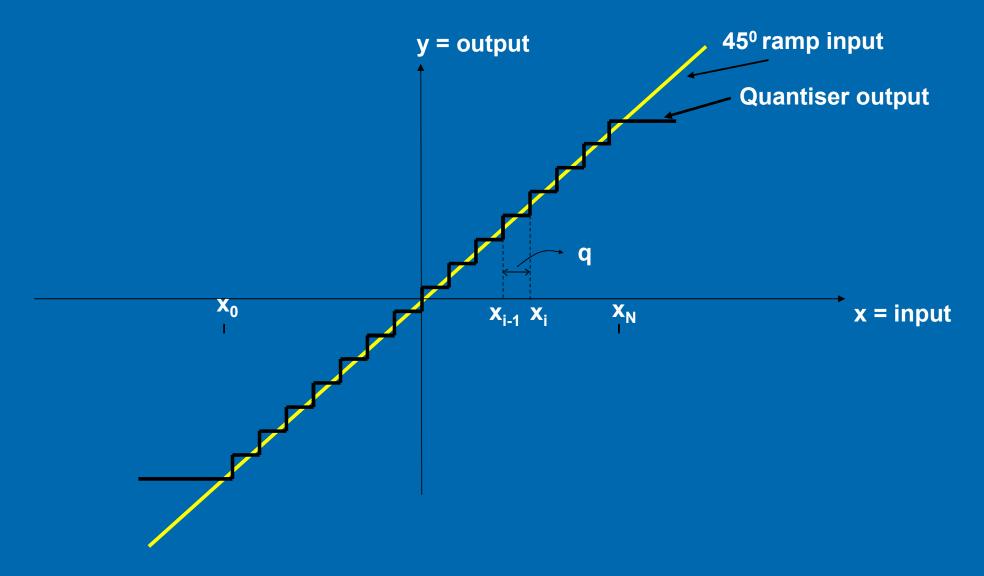
### **Overload Error**

Besides quantisation error or quantising noise the quantisation process also causes overload error or clipping.





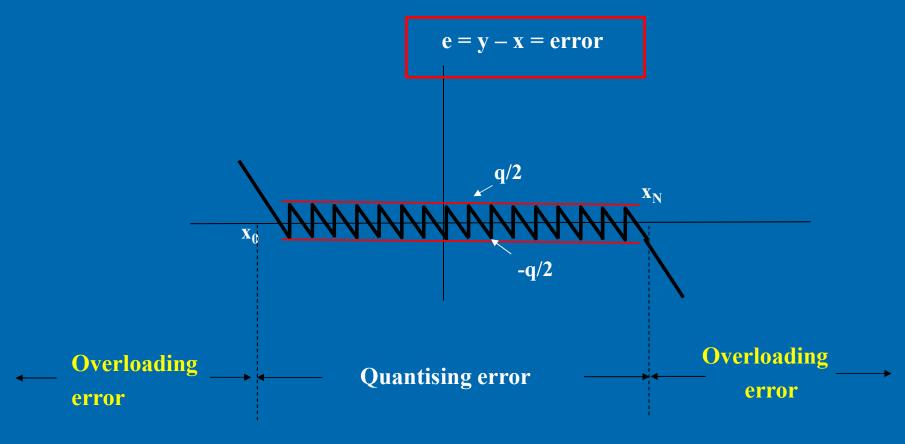
## **Quantisation noise power**





### **Quantisation noise power**

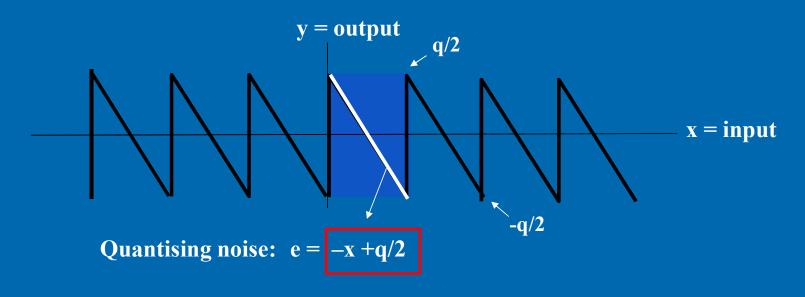
 The difference between the quantised output and input waveform gives the quantisation noise (waveform), denoted as e.





### **Quantisation noise power**

### **Quantisation Noise Waveform**



rms value of the quantisation noise signal:

$$e_{\rm rms} = \sqrt{\frac{1}{q} \int_0^q \left(-x + \frac{q}{2}\right)^2 dx} = \sqrt{\frac{q^2}{12}}$$



### **Quantisation noise power**

• The quantisation noise power (over a  $1\Omega$  load) is

$$N_q = \frac{e_{rms}^2}{R} = e_{rms}^2$$
 (as R = 1)

$$N_q = \frac{q^2}{12}$$
 watts  $(R=1)$ 

- The result is applicable to any input to an uniform quantiser.
- The same result can be obtained when the quantisation noise is a random signal with an uniform distribution in the interval -q/2 to +q/2, i.e.

$$p_e(e) = \frac{1}{q},$$
  $-\frac{q}{2} \le e \le +\frac{q}{2}$ 

$$= 0, \quad \text{otherwise}$$



### Signal to quantising noise (S/N<sub>q</sub>)

The performance of a quantiser is measured by signal-to-noise ratio that takes both quantising error and overload error into account.

$$SNR = \frac{S}{N_o + N_q} \approx \frac{S}{N_q}$$
 Overload error is neglected for simplicity.

- Quite often SNR calculation for a quantiser is based on sinusoidal inputs because SNR results for speech and sinusoidal inputs are quite close.
- Use of sinusoidal make the measurements and calculation of  $S/N_{\rm q}$  easier.



### Signal to quantising noise (S/N<sub>q</sub>)

### **Derive formula for S/N**<sub>q</sub>

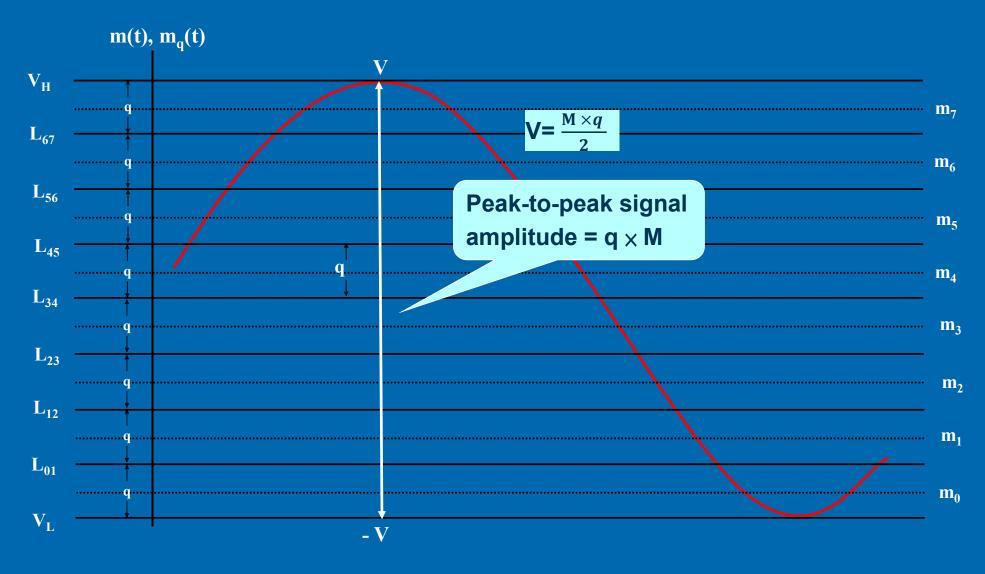
 For a full range (-V to +V) sinusoidal input that has zero overload error, the average signal power is

$$S = \frac{V_{rms}^2}{R} \qquad V_{rms} = \frac{V}{\sqrt{2}} \qquad \text{V = peak signal amplitude}$$

$$S = \frac{V^2}{2} \qquad (R = 1)$$

Peak-to-peak signal amplitude = input range of quantiser







# Signal to quantising noise (S/N<sub>g</sub>) Derive formula for S/N<sub>g</sub>

For a full range (-V to +V) sinusoidal input that has zero overload error, the average signal power is

$$S = \frac{V_{rms}^2}{R} = V_{rms}^2 \qquad (R = 1)$$

$$S = \frac{V^2}{2}$$
 Since  $V_{rms} = \frac{V}{\sqrt{2}}$ 

The peak value V of the sinusoid can be expressed in terms of step size q and number of levels in the quantiser M, as follows:

$$2V = qM$$

$$V = \frac{qM}{2}$$

$$\therefore rms \ value = V_{rms} = \frac{V}{\sqrt{2}} = \frac{qM}{2\sqrt{2}}$$



### Signal to quantising noise (S/N<sub>g</sub>) Derive formula for S/N<sub>g</sub>

Hence, the average signal power is

$$S = V_{rms}^2 = \left(\frac{qM}{2\sqrt{2}}\right)^2 = \frac{q^2M^2}{8}$$

Combining eqs for S and N<sub>q</sub>

$$N_q = \frac{q^2}{12}$$

$$\frac{S}{N_q} = \frac{q^2 M^2}{8} \cdot \frac{12}{q^2} = 1.5 M^2$$

$$\left[ \frac{S}{N_q} \right] = 10 \log_{10} (1.5M^2) = 10 \log_{10} (1.5) + 10 \log_{10} (M^2)$$

$$\left[ \frac{S}{N_q} \right]_{dB} = 1.76 + 20 \log_{10} M$$



# Signal to quantising noise (S/N<sub>q</sub>) Derive formula for S/N<sub>q</sub>

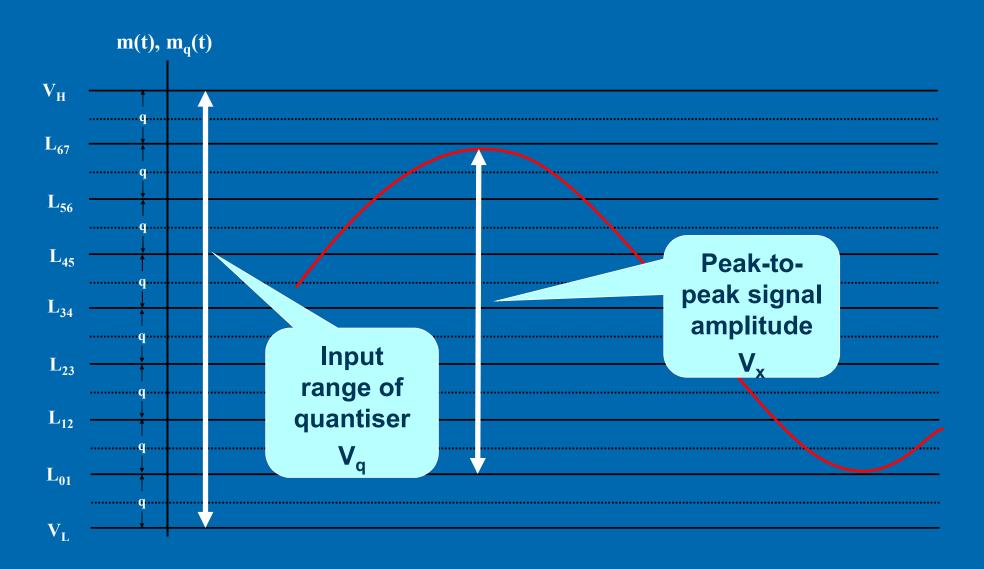
Since 
$$M = 2^B$$
,

$$\left[\frac{S}{N_q}\right]_{dB} = 1.76 + 20log_{10}(2^B) = 1.76 + B \times 20log_{10}(2)$$

$$\left[\frac{S}{N_q}\right]_{dB} = 1.76 + 6B \text{ dB}$$

 $\begin{bmatrix} \frac{S}{N_q} \end{bmatrix}_{dR}$  = 1.76 + 6B dB for a sinusoid whose amplitude range coincides with the range of the quantiser. with the range of the quantiser.







### Signal to quantising noise (S/N<sub>q</sub>)

For sinusoidal inputs whose amplitude, V<sub>x</sub>, is less than the full input range of the quantiser, V, then

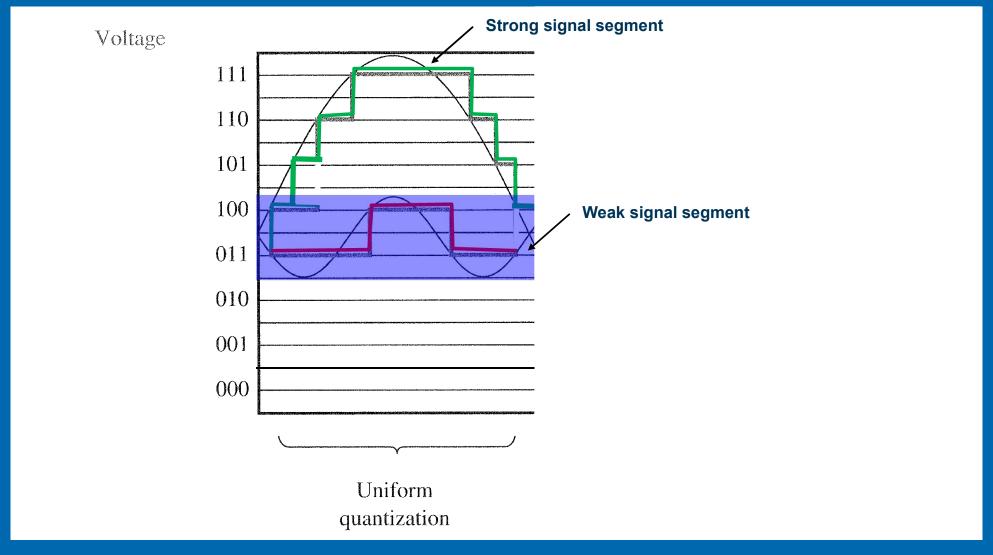


### Non-uniform quantization

- For signals with large variance in strength over time, it is preferable to use a non-uniform quantiser i.e. a quantiser that has variable step size.
  - E.g. Speech signal strength can vary largely from one speech segment to another.
    - If the quantiser is designed to accommodate strong signals (like vowels), the quantisation step size will be large.
    - Weaker signals (like consonants) will be subjected to a larger quantisation error.

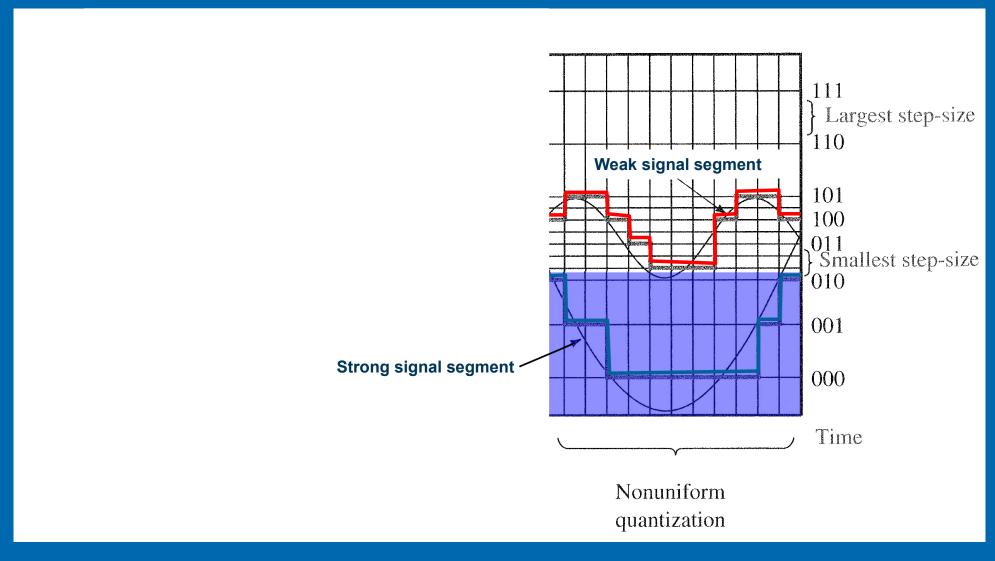
- The quality of the speech will be affected.





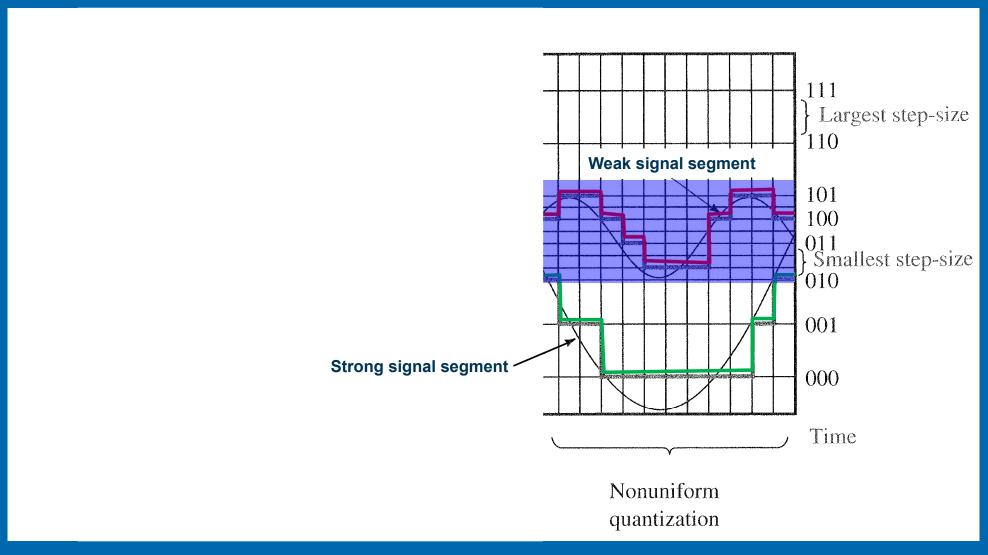




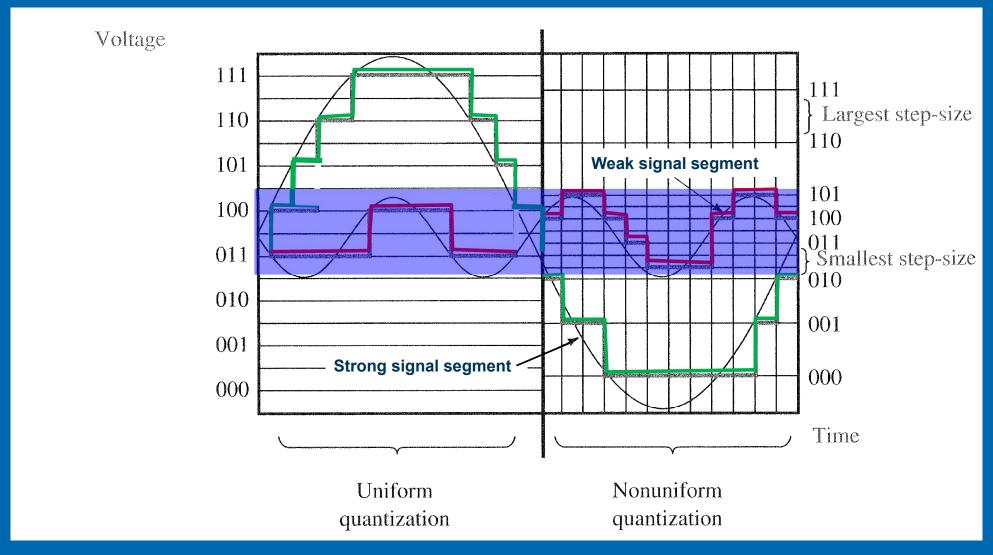










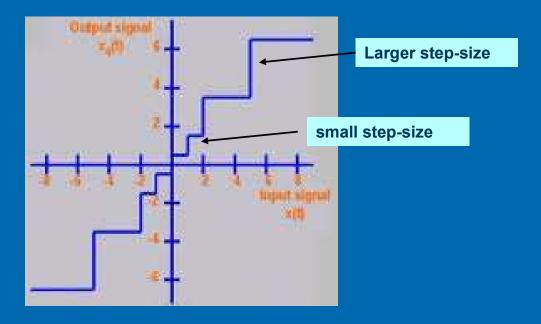




### **Non-uniform quantization**

 For signals with large varying amplitude, a suitable non-uniform quantiser would be a quantiser whose step size increases with the signal amplitude.

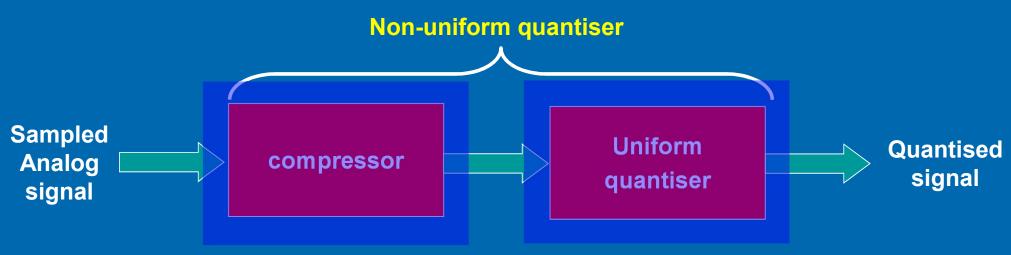
### I/O characteristic of a non-uniform quantiser





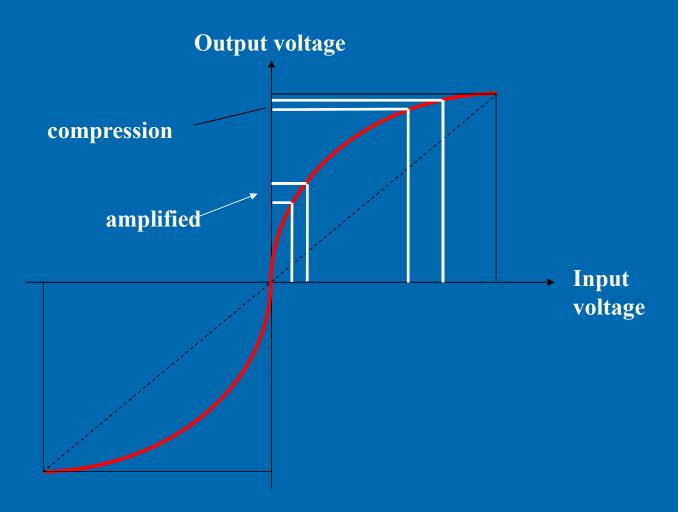


- In practice, a non-uniform quantizer is implemented by combining a uniform quantizer with a compressor.
- A sampled analog signal is first input to a compressor and then to a uniform quantiser.
- The compressor can be viewed as a variable-gain amplifier that amplifiers the signal at low amplitude and attenuate the signal at high amplitude.
- The compressor and uniform quantiser work jointly to form a non-uniform quantiser.





### **Non-uniform quantization**



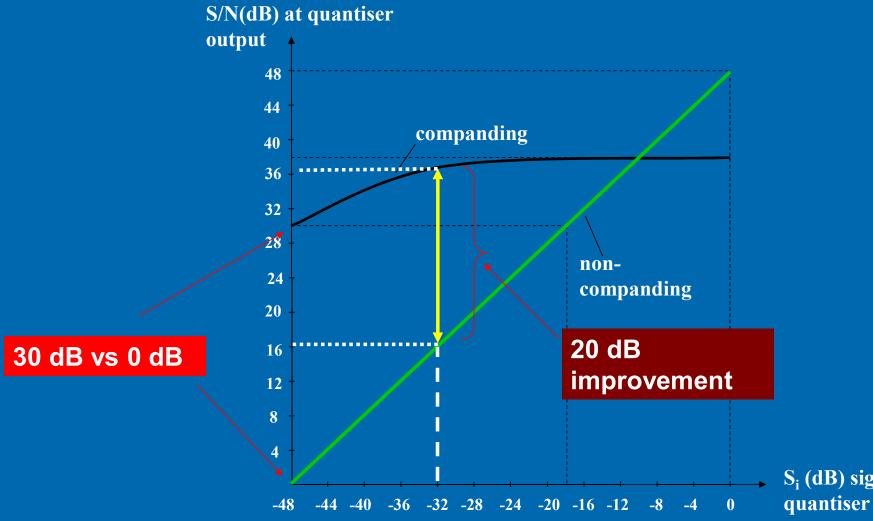
Input-output characteristic of a compressor





- At the receiver the signal is reconstructed by the reverse process.
   i.e. by expanding it.
- This process of compression-expansion is called COMPANDING.
   (COMpressing-exPANDING)





 $S_i$  (dB) signal power at quantiser input



# **END**

**CHAPTER 7** 

(Part 2 of 4)

