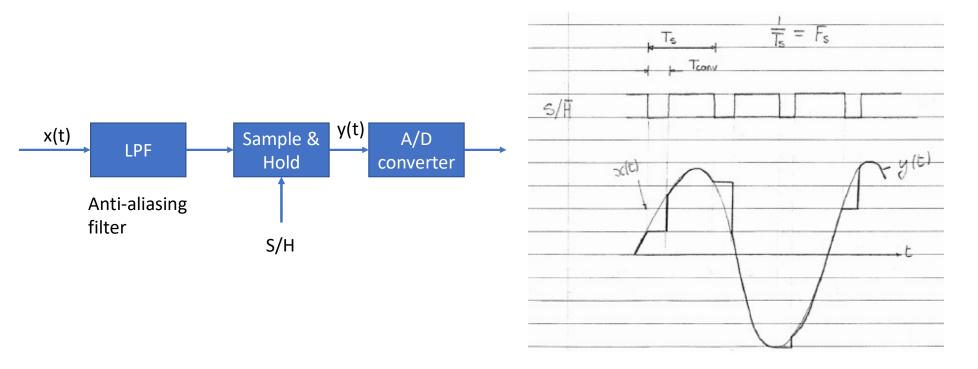


Quantisation (see Stremler section 9.4)





Quantisation

• The incoming signal is quantised to a certain number of voltage levels, and each voltage level is assigned a unique binary code by the ADC $No.\ of\ levels=2^B$

B = number of bits

e.g.
$$B = 8 \Rightarrow 256$$
 levels

• Quantisation step size $\Delta = \frac{2A}{2^B - 1}$

Where A is the amplitude of a full scales sinusoid



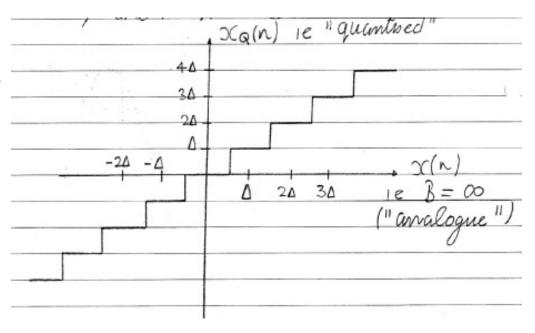
Quantisation - example

- If an ADC has an input range of $\pm 5V$, then A=5V
- $x_Q(n)$ takes on a value of $k\Delta$, if the input voltage is between $k\Delta-\frac{\Delta}{2}$ and $k\Delta+\frac{\Delta}{2}$
- i.e. the "quantisation error" is

$$e(n) = \left| x(n) - x_Q(n) \right| \le \frac{A}{2}$$

Or

$$-\frac{\Delta}{2} \le e(n) \le \frac{\Delta}{2}$$



Quantisation noise

• Assuming that the errors are equally likely to occur anywhere in this range, we can write a probability density function for e(n) as

$$P[e(n)] = \frac{1}{\Delta}, \qquad -\frac{\Delta}{2} \le e(n) \le \frac{\Delta}{2}$$
$$\overline{e(n)} = 0$$

Variance:

$$\overline{e^2(n)} = \int_{-\infty}^{\infty} p[e]e^2 de - \overline{e(n)}$$

$$\int_{-\frac{\Delta}{2}}^{\frac{\Delta}{2}} \frac{1}{\Delta} e^2 de = \frac{\Delta^2}{12}$$

$$\Delta = \frac{2A}{2^B - 1} \Rightarrow \overline{e^2(n)} = \frac{4A^2}{12(2^B - 1)^2}$$

• Mean squared value of a full scale sinusoid is

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

 We can use this to calculate the Signal to Quantization Noise ratio...

Signal to Quantisation Noise

"Signal to Quantisation Noise" ratio is

$$SQNR = \frac{\frac{A^2}{2}}{\frac{4A^2}{12(2^B - 1)^2}}$$
$$= \frac{12(2^B - 1)^2}{2.4}$$
$$= \frac{3}{2}(2^B - 1)^2$$

• Take logs to get

$$SQNR_{dB} = 10 \log_{10} \frac{3}{2} + 10 \log_{10} (2^{B} - 1)^{2}$$

$$= 10 \log_{10} \frac{3}{2} + 20 \log_{10} (2^{B} - 1)$$

$$\approx 10 \log_{10} \frac{3}{2} + 20 \log_{10} 2^{B}$$

$$= 6B + 1.8dB$$

e.g. 12 bit A/D converter has a theoretical SQNR of 73.7dB



Pulse Code Modulation (PCM)

- Samples of analogue signal are converted into binary numbers
- Widely used in telephony, where each binary number representing a sample is converted into serial form, and transmitted along a telephone line

$$Bit\ rate = B \times F_s$$

B-no. of bits/sample $F_s-sampling rate$

- Telephony: $F_s = 8kHz$
- Where Δ is fixed, we have linear or uniform PCM
- Problem: SQNR decreases with decreasing signal level



PCM – Problems with speech

Problems

- i) power is quite low compared to peak value (modulation index)
- ii) variations in volumes between speakers
- If linear uniform PCM is used, SQNR varies depending on volume

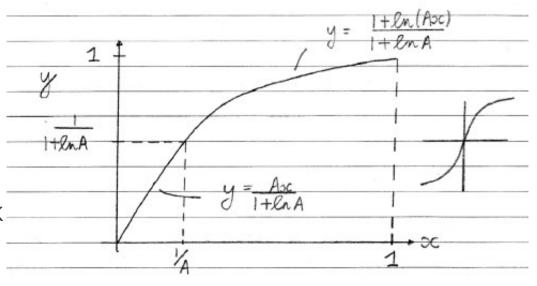
Solution

- Non-uniform quantisation i.e. small step sizes for low amplitudes, larger step sizes for larger amplitudes
- Result: SQNR is roughly constant over entire range of signal



Log PCM

- Two variants:
 - A-Law (Europe)
 - μ-Law (USA)
- Note: x and y are normalised
- Typical values: A=87.6, B=8,Fs=8k
- ⇒64kbits/s "toll quality"
- Equivalent to 12 bit uniform PCM





Practical Implementation

- In practice, a piecewise linear approximation is used for the compression equations
- Positive part: 8 linear segments
- Note: compression→expanding = "companding"

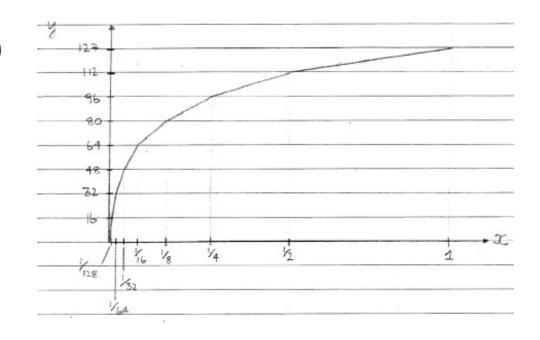


Practical Implementation

- First segment: codes 0 to 15 assigned uniformly to sample values between 0 and 1/128 of the max positive input
 - ⇒ equivalent to 12-bit linear PCM
- Second segment: codes 16 to 31 assigned to voltage between 1/128 and 1/64 of max input
 - ⇒ equivalent to 12-bit linear PCM

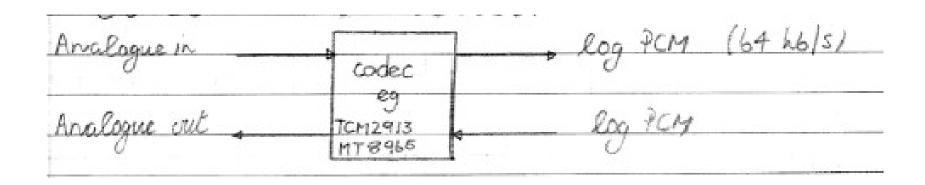
Etc...

- Codes 112 to 127 assigned uniformly to inputs between ½ and 1
 - ⇒ 6-bit linear PCM





"Codec" – coder/decoder





SQNR for arbitrary Signals

- Previously, we calculated the SQNR for a linear ADC, for the case of a signal consisting of a single sinewave (about as simple as it gets)
- We then looked at specific techniques for non-uniform quantisation of speech
- Here, we revisit uniform quantisation, but without placing any constraints on the nature of the signal
- We are interested in
 - Determining the SQNR for a given number of bits
 - Determining the required number of bits for a desired SQNR



SQNR for arbitrary Signals - analysis

For a given number of bits, the SQNR is simply

$$SQNR = (V_{RMS})^2 / \overline{e^2(n)}$$

Where V_{RMS} is the signal RMS value, and $\overline{e^2(n)}$ is the mean squared value of the quantisation noise, and is equal to $\frac{\Delta^2}{12}$

- Example:
 - Calculate SQNR for an ADC with B=12bits, an input voltage range of ±3V, and for an input signal with a peak value of 2.4V and a peak-to-RMS ratio (PRR) of 1.9



SQNR for arbitrary Signals - analysis

$$V_{RMS} = \frac{V_{peak}}{PRR} = \frac{2.4}{1.9} = 1.263V$$

$$\Delta = 2 \times \frac{3}{2^{12} - 1} = 1.46mV$$

$$\Rightarrow \overline{e^2(n)} = \frac{\Delta^2}{12} = 0.178\mu V^2$$

$$\Rightarrow \text{SQNR} = \frac{(1.263)^2}{0.178 \times 10^{-6}} = 8.962 \times 10^6$$

$$= 69.5dB$$



SQNR for arbitrary Signals - design

- To calculate the number of bits required to achieve a certain minimum SQNR, we basically work backwards
- i.e. we need to determine the maximum permissible value of $e^2(n)$, then determine Δ , and hence determine B



Example - design

- Example:
 - V_{peak} =2.3V, PRR=3.9, desired SQNR \geq 70dB
 - ADC input range = ± 2.5 (i.e. A=2.5)
- Step 1: Determine V_{RMS}

$$V_{RMS} = \frac{2.3}{3.9} = 0.59V$$

• Step 2: from V_{RMS} , and SQNR, determine $\overline{e^2(n)}$

$$SQNR = \frac{(V_{RMS})^2}{\overline{e^2(n)}} \ge 10^7 (i.e.70dB)$$

$$\Rightarrow \frac{(0.59)^2}{e^2(n)} \ge 10^7$$

$$\overline{e^2(n)} \le 0.0348 \times 10^{-6}$$



SQNR for arbitrary signals

• Step 3. Given $\overline{e^2(n)}$, determine upper limit on Δ

$$\overline{e^2(n)} = \frac{\Delta^2}{12} \le 0.0348 \times 10^{-6}$$

 $\Rightarrow \Delta \le 0.646 mV$

• Step 4. From Δ and the ADC input voltage range, determine B

$$\Delta = \frac{2A}{(2^B - 1)} \le 0.646 mV$$

$$\Rightarrow 2^B - 1 \ge \frac{2(2.5)}{0.646 \times 10^{-3}}$$

Solve to get $B \ge 12.91 \ i.e. B = 13$

