Typical stages in digital signal processing A/D Conversion

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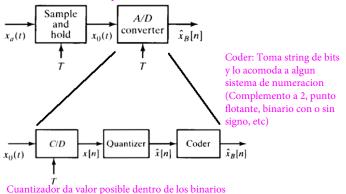
Summary

- A/D Conversion Stages
- Quantizer
- 3 ADC Signal-to-Noise relationship
- 4 ADC resolution for a particular signal

A/D Conversion Stages

- The A/D converter is a physical device that converts a voltage or current amplitude at its input into a binary code representing a quantized amplitude value closest to the amplitude of the input.
- The sample-and-hold stage can be a zero-order-hold.

S&H mantiene un valor de tensión durante un tiempo determinado a la entrada del A/D, ya que la conversión de A/D toma un tiempo finito.



Quantizer

- Uniformly spaced quantizer, function Q(x).
- The number of quantization levels will be a power of two (2^B).
- Precision of quantizer, $\Delta = \frac{\text{full voltage range}}{2^{\text{word lenght}}} = \frac{2V_p}{2^B}$. [mV]

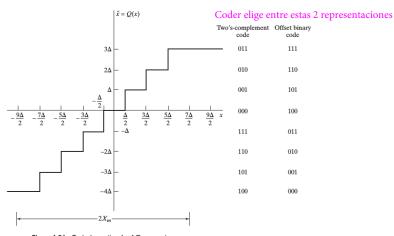
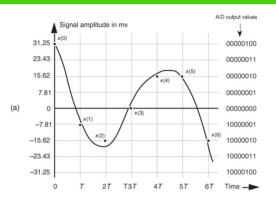
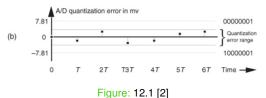


Figure 4.54 Typical quantizer for A/D conversion.

Quantizer Error example





Quantizer Error example, II

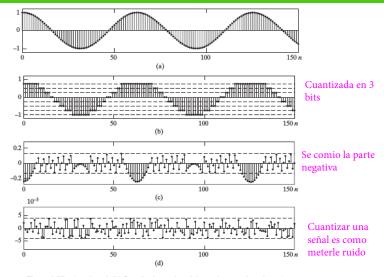


Figure 4.57 (continued) (b) Quantized samples of the cosine waveform in part (a) with a 3-bit quantizer. (c) Quantization error sequence for 3-bit quantization of the signal in (a). (d) Quantization error sequence for 8-bit quantization of the signal in (a).

Quantizer Error model

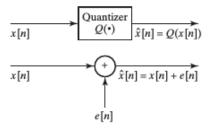


Figure 4.56 Additive noise model for quantizer.

Cuantizador "empeora" señal original al agregarle ruido

ADC Signal-to-Noise relationship

The precision of the quantizer is given by [2]:

$$\Delta = q = \frac{\text{full voltage range}}{2^{\text{word lenght}}} = \frac{2V_p}{2^B}. \quad [mV]$$
 (1)

Relacion señal a ruido

- $SNR = (P_{signal})/(P_{noise})$ relates two powers. potencia
- Since q is defined as a random variable, its power cannot be represented explicitly.
- A statistical version of SNR is used.

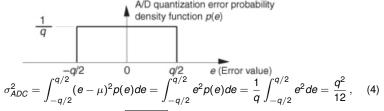
$$SNR_{ADC} = 10 \cdot \log_{10} \left(\frac{\text{input signal variance}}{A/D \text{ quantization noise variance}} \right), \quad [dB]$$
 (2)

$$= 10 \cdot \log_{10} \left(\frac{\sigma_{signal}^2}{\sigma_{ADC}^2} \right) . \tag{3}$$

Relaciona varianzas en vez de potencias

En valores de

ADC Signal-to-Noise relationship, II



$$\sigma_{ADC}^2 = \left(\frac{2V_p}{2^B}\right)^2 \cdot \frac{1}{12} = \boxed{\frac{V_p^2}{3 \cdot 2^{2B}}},$$
 (5)

Load Factor,
$$LF = \frac{rms_{signal}}{V_p} = \frac{\sigma_{signal}}{V_p} \implies \sigma_{signal}^2 = \boxed{LF^2 \cdot V_p^2},$$
 (6)

$$\begin{split} \textit{SNR}_{ADC} &= 10 \cdot \log_{10} \left(\frac{\sigma_{signal}^2}{\sigma_{ADC}^2} \right) \;, \\ &= 10 \cdot \log_{10} \left[\left(\textit{LF}^2 \cdot \textit{V}_p^2 \right) \cdot \frac{3 \cdot 2^{2B}}{\textit{V}_2^2} \right] = 10 \cdot \log_{10} \left[\left(\textit{LF}^2 \cdot 3 \cdot 2^{2B} \right) \right] \;, \end{split}$$

$$= 10 \cdot \left[\log_{10}(LF^2) + \log_{10}(3) + 2\log_{10}(2) \cdot B \right] , \tag{8}$$

$$= 20 \cdot \log_{10}(LF) + 4.77 + 6.02 \cdot B. \quad [dB]$$
 (9)

(7)

ADC Signal-to-Noise relationship considerations

$$SNR_{ADC} = 20 \cdot \log_{10}(LF) + 4.77 + 6.02 \cdot B$$
, [dB]
= $20 \cdot \log_{10}\left(\frac{rms_{signal}}{V_p}\right) + 4.77 + 6.02 \cdot B$. [dB]

Se busca SNR lo más grande posible->Aumentar cantidad de Bits, o el LF.

Considerations about LF:

- Ideally, if rms_{signal} >> V_p, SNR_{ADC} increases, but this will produce a severe distortion in the sampling signal (saturation).
- On the other hand, if $rms_{signal} << V_p$, SNR_{ADC} decreases. busca mucho mas grande que AD es muy mala idea alimentarlo con miliVolts. tension pico. Pero hay limites

Considerations about *B* (numbers of bits):

- SNR_{ADC} increases 6 dB by each bit in ADC's quantizer.
- So, the more bits the better, isn't it?

Potencia efectiva de la señal se busca mucho mas grande que tension pico. Pero hay limites físicos. A un ADC no podes darle tanta potencia efectiva, porque son de baja

Suelen ir de la mano rms y Vp

Other sources of error should be taken into account:

- It was considered that ADC's $V_{MAX} = V_p$. No trabajamos en picos constantemente. Silencio en audio por ej
- The proposed model for tje probability density function of e[n] may not be uniform.
- Therefore, SNR_{ADC} from Eq. 9 should be decreased by 3 or 6 dB.
 Esto es una pérdida por consideraciones no tenidas en cuenta

ADC Signal-to-Noise relationship for a sinusoidal signal

For a sinusoidal signal, $rms_{signal} = V_p/\sqrt{2}$.

$$SNR_{ADC} = 20 \cdot \log_{10} \left(\frac{rms_{signal}}{V_{\rho}} \right) + 4.77 + 6.02 \cdot B,$$

$$= 20 \cdot \log_{10} \left(\frac{V_{\rho} / \sqrt{2}}{V_{\rho}} \right) + 4.77 + 6.02 \cdot B.$$
(10)

Thus, the maximum SNR_{ADC} is,

$$SNR_{ADC} = 20 \cdot \log_{10} \left(1/\sqrt{2} \right) + 4.77 + 6.02 \cdot B,$$
 (12)

$$= -3.01 + 4.77 + 6.02 \cdot B, \tag{13}$$

$$= 1.76 + 6.02 \cdot B. \text{ [dB]} \tag{14}$$

LF hace aporte negativo.

ADC resolution for a particular signal

Consider the following example:

- The SNR for an audio output amplifier is 110 dB.
- A 24-bits ADC is chosen to sample the output amplifier (professional audio).
- The amplifier is excited by an input sinusoidal signal. -3 es factor de seguridad por consideraciones
 SNR_{ADC} = 1.76 + 6.02 · 24 3 = 143.24 dB.

- How many bits are used to measure noise? $(143 110)/6 \simeq 5.5$ bits!.
- In a control loop, picking a bad ADC resolution could lead to a catastrophic scenario.
- What happens if B=10 bits? $SNR_{ADC}=61.96$ dB, $(110-61.96)/6 \simeq 8$ extra bits are needed!.

Summary: the number of bits B in an ADC must provide $SNR_{ADC} \geqslant SNR_{signal}$.

- Rule of thumb: the ADC resolution should be choose in order to provide 6 dB (1 bit) above the SNR of the signal to be sample.
- Additional bits (noisy bits) can be eliminated by right shifting.

Bibliography

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- 3 Paolo Prandoni and Martin Vetterli. Signal processing for communications. Taylor and Francis Group, LLC. 2008. Section 9.6.