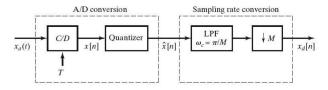
Digital Signal Processing Oversampled Analog to Digital Conversion with Noise Shaping

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Oversampled ADC System Model

Oversampling with direct quantization model:



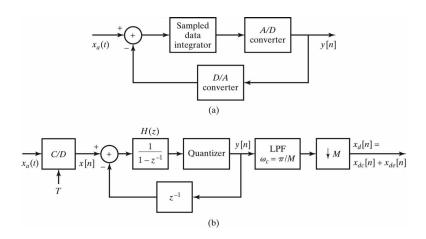
We've previously calculated the SNR of this system as

$$SNR_Q = 6.02B + 10.8 - 20\log_{10}\left(\frac{X_m}{\sigma_x}\right) + 10\log_{10}(M)$$

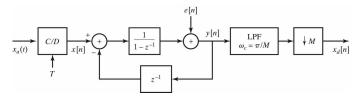
To achieve 60 dB SNR without oversampling, we need B=10 bits. Oversampling by a factor of M=256 reduces the number of necessary bits to B=6.

To achieve more gains, we can combine oversampling with "noise shaping". The main idea is to modify the A/D conversion procedure so that the power spectrum of the quantization noise is pushed outside the passband of the LPF.

Oversampled Quantizer with Noise Shaping System



Additive Quantization Noise Model (part 1 of 2)



Since everything is linear, we can analyze the transfer function between x[n] and y[n] separately from the transfer function from e[n] to y[n]. We have

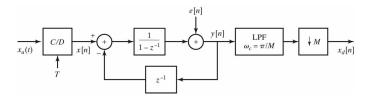
$$Y(z) = \frac{1}{1 - z^{-1}} \left(X(z) - z^{-1} Y(z) \right)$$

which can be rearranged as

$$\frac{1}{1-z^{-1}}Y(z) = \frac{1}{1-z^{-1}}X(z)$$

hence $H_x(z) = 1$.

Additive Quantization Noise Model (part 2 of 2)



Looking now at the quantization noise, we have

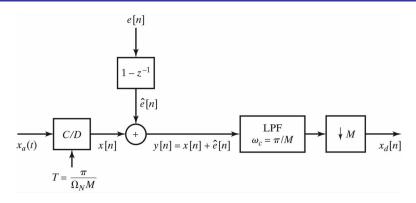
$$Y(z) = \frac{z^{-1}}{1 - z^{-1}}Y(z) + E(z)$$

which can be rearranged as

$$\frac{1}{1 - z^{-1}} Y(z) = X(z)$$

hence $H_e(z) = 1 - z^{-1}$.

Equivalent Model

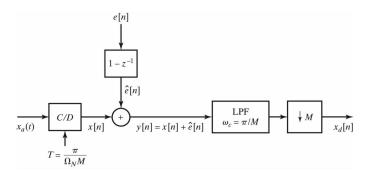


Recall

$$x_d[n] = \underbrace{x_{da}[n]}_{\text{signal part}} + \underbrace{x_{de}[n]}_{\text{noise part}}$$

We already know from previous analysis that the power of the signal part is unchanged by such a system, i.e., $\sigma_{x_{da}}^2 = \sigma_x^2 = \sigma_{x_a}^2$.

Noise Power Analysis (part 1 of 2)



Since
$$\Phi_{ee}(e^{j\omega}) = \sigma_e^2$$
 for all ω , we have

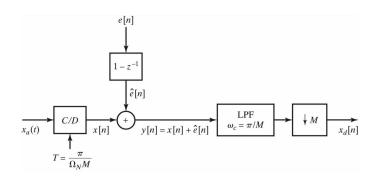
$$\Phi_{\hat{e}\hat{e}}(e^{j\omega}) = |H_e(e^{j\omega})|^2 \sigma_e^2 = |1 - e^{-j\omega}|^2 \sigma_e^2 = 4\sin^2(\omega/2)\sigma_e^2.$$

Note that

$$\frac{1}{2\pi} \int_{-\pi}^{\pi} 4\sin^2(\omega/2) \sigma_e^2 d\omega = 2\sigma_e^2$$

hence the total quantization noise power has doubled after the filter.

Noise Power Analysis (part 2 of 2)

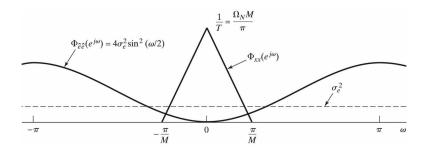


The **actual** quantization noise power that gets through to $x_d[n]$ however is

$$\sigma_{x_{de}}^2 = \frac{1}{2\pi} \int_{-\pi/M}^{\pi/M} 4 \sin^2(\omega/2) \sigma_e^2 \, d\omega = 2 \left(\frac{1}{M} - \frac{\sin(\pi/M)}{\pi}\right) \sigma_e^2 \approx 2 \left(\frac{\pi^3/M^3}{6\pi}\right) \sigma_e^2$$

where the approximation results from $\sin(x) \approx x - \frac{x^3}{3!}$ for small x.

Oversampled Quantizer with Noise Shaping System



SNR Analysis

Recall that a B+1 bit uniform quantizer with step size $\Delta=\frac{X_m}{2^B}$ has $\sigma_e^2=\frac{\Delta^2}{12}=\frac{X_m^2}{12\cdot 2^{2B}}.$ Hence, the SNR of an oversampled ADC with noise shaping is

$$\begin{aligned} \text{SNR}_Q &= 10 \log_{10} \left(\frac{\sigma_{x_{da}}^2}{\sigma_{x_{de}}^2} \right) \\ &= 10 \log_{10} \left(\frac{\sigma_{x_{de}}^2}{2 \left(\frac{\pi^2 / M^3}{6} \right) \sigma_e^2} \right) \\ &= 10 \log_{10} \left(\frac{12 \cdot 2^{2B} \sigma_x^2 \cdot 3M^3}{\pi^2 X_m^2} \right) \\ &= 6.02B + 5.62 - 20 \log_{10} \left(\frac{X_m}{\sigma_x} \right) + 30 \log_{10}(M) \end{aligned}$$

Remarks:

- ▶ Oversample and shape the quantization noise away from the signal band.
- lacktriangle Doubling M and keeping B fixed gives a 9 dB SNR improvement.
- lackbox Quadrupling M and using two less quantizer bits gives the same SNR.

