

Instituto Superior Técnico

Sistemas de Processamento Digital de Sinais

Sigma-Delta A/D converter

Consider a first order $\Sigma\Delta$ A/D converter for which the noise modulator is as depicted in figure 1a). The output 1-bit signal $y(n)$ is filtered by an ideal digital lowpass filter with transfer function $H(z)$ with magnitude represented in figure 1b) and then decimated by a factor M .

- Determine the maximum signal-to-noise ratio (SNR) at the output $z(n)$ and plot its variation in dB as a function of the oversampling factor M .
- Determine and plot the effective number of bits as a function of M .
- Explain how many dB by octave of M are achieved by oversampling and by noise-shaping.

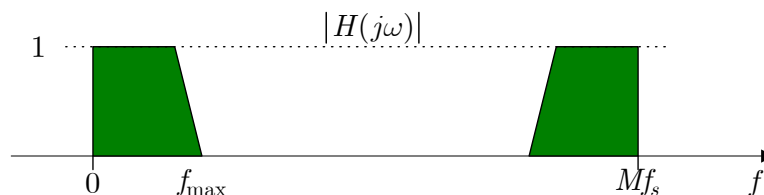
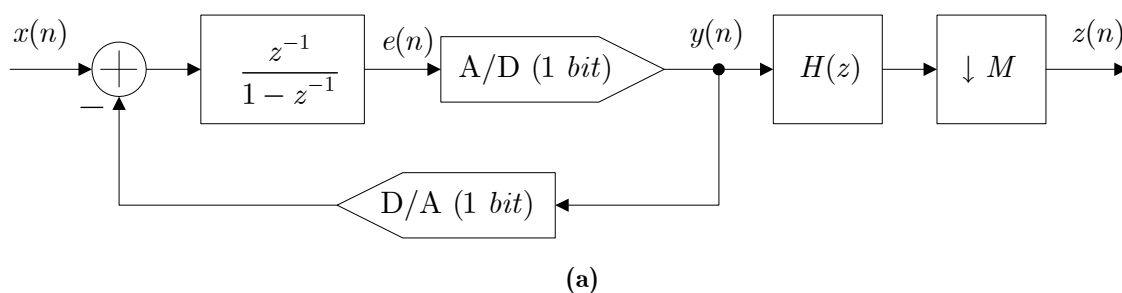
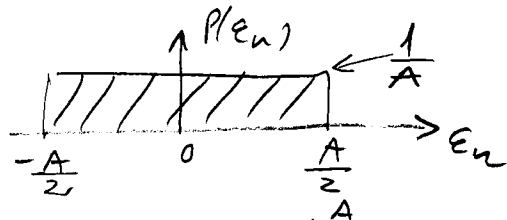
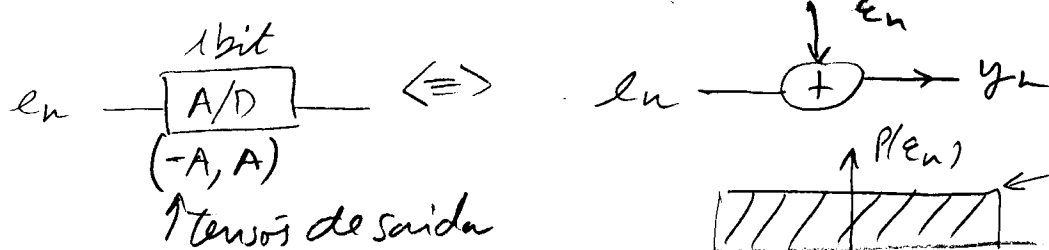


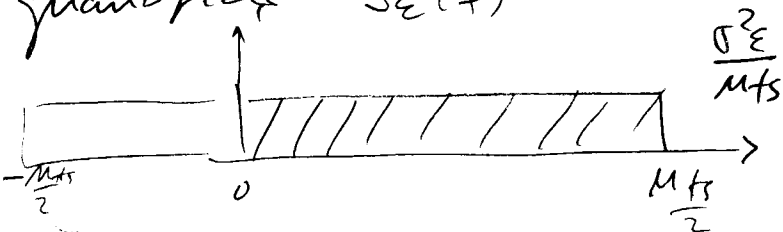
Figure 1: a) Sigma-Delta noise shaping model; b) Decimator magnitude filter response.



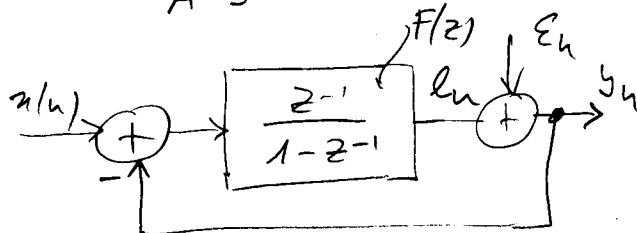
$f_s = 2f_{max} \rightarrow$ Nyquist

$Mfs \rightarrow$ freq. a que o conversor funciona.

Densidade espectral de ruído de quantificação $S_E(f)$



$$\sigma_{e_n}^2 = \int_{-A}^A e^2 p(e) de = \frac{1}{A} \int_{-A/2}^{A/2} e^2 de = \frac{1}{A} \left. \frac{e^3}{3} \right|_{-A/2}^{A/2} = \frac{1}{A} \frac{1}{3} \left(\frac{A}{2} \right)^3 \cdot 2 = A^2/12$$

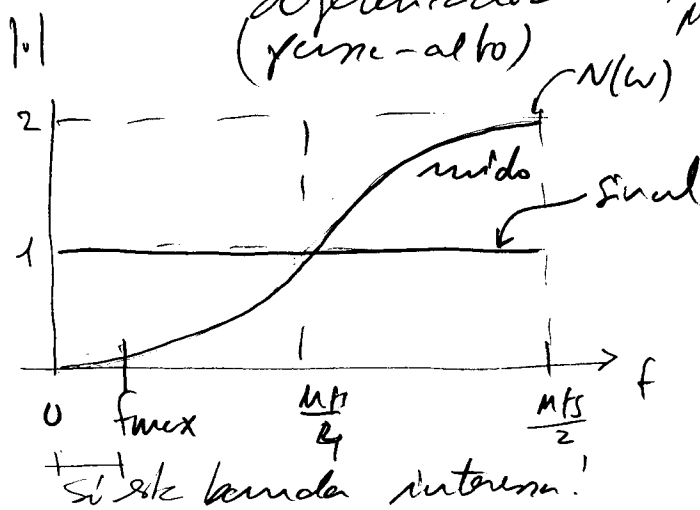


$$Y(z) = E(z) + [X(z) - Y(z)]F(z)$$

$$Y(z)[1 + F(z)] = E(z) + F(z)X(z)$$

$$Y(z) = \frac{1}{1 + F(z)} E(z) + \frac{F(z)}{1 + F(z)} X(z)$$

$$= \underbrace{(1 - z^{-1})}_{\text{diferenciador (passe-alto)}} E(z) + \underbrace{z^{-1}}_{\text{atraso de uma amostra}} Y(z)$$



$$|z^{-1}|_{z=e^{j\omega T_s}} = 1$$

$$|1 - z^{-1}|_{z=e^{j\omega T_s}} =$$

$$= |1 - e^{-j\omega T_s}| =$$

$$= |2 \sin \frac{\omega T_s}{2}| = N(\omega)$$

$$= |1|^2 \times 2\pi f \frac{T_s}{2M} = \frac{\pi f T_s}{M}$$

Potência de ruído na banda.

$$\sigma_{E_{banda}}^2 = \int_{-f_{max}}^{f_{max}} S_E(f) df$$

$$S_E(f) df = 2 \frac{\sigma_E^2}{Mfs} \left(2 \sin \frac{2\pi f T_s}{2M} \right)^2 df$$

$$\approx \frac{8 \sigma_E^2}{Mfs} \int_{-f_{max}}^{f_{max}} \left(\frac{\pi f T_s}{M} \right)^2 df = \frac{8 \pi^2 T_s^2}{Mfs M^2} \frac{f^3}{3} \Big|_{-f_{max}}^{f_{max}} = \frac{8 \pi^2 T_s^2}{Mfs M^2} \frac{f_{max}^3}{3} \sigma_E^2$$

$$f_s = 2 f_{max} \rightarrow f_{max} = \frac{f_s}{2}$$

$$\sigma_{\epsilon}^2_{band} = 8\pi^2 \frac{f_s^2}{M^2} \sigma_{\epsilon}^2 \left(\frac{f_s}{2}\right)^3 \times \frac{1}{3Mf_s} = \frac{\pi^2}{3M^3} \times \sigma_{\epsilon}^2 \rightarrow \frac{\pi^2}{3M^3} \sigma_{\epsilon}^2$$

$$= \frac{\pi^2}{3M^3} \frac{A^2}{12} = \frac{\pi^2}{36M^3} A^2$$

$$Pot. de sinal = A^2/2$$

$$SNR = \frac{A^2/2}{\frac{\pi^2}{36M^3} A^2} = \frac{18}{\pi^2} M^3$$

$$SNR_{dB} = 10 \log_{10} \frac{18M^3}{\pi^2} = 30 \log_{10} M + 10 \log_{10} \left(\frac{18}{\pi^2} \right)$$

2.61 dB

$$30 \log_{10} M = 30 \log_{10} 2 \log_2 M$$

$$\approx 9 \log_2 M$$

9 dB/octava

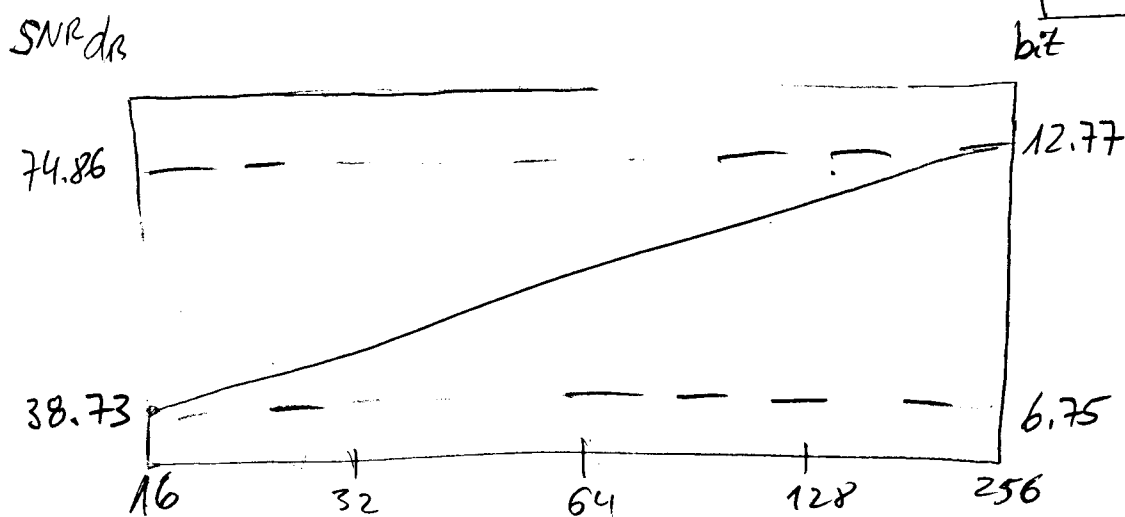
3 dB (oversampling)
6 dB (filtro)

$$SNR = 64 - 1.76 \text{ dB}$$

$$= 9 \log_2 M + 2.61$$

$$w_b = \frac{9 \log_2 M + 4.37}{6}$$

$$= \frac{3}{2} \log_2 M + 0.73$$



$M \equiv$ factor de oversampling