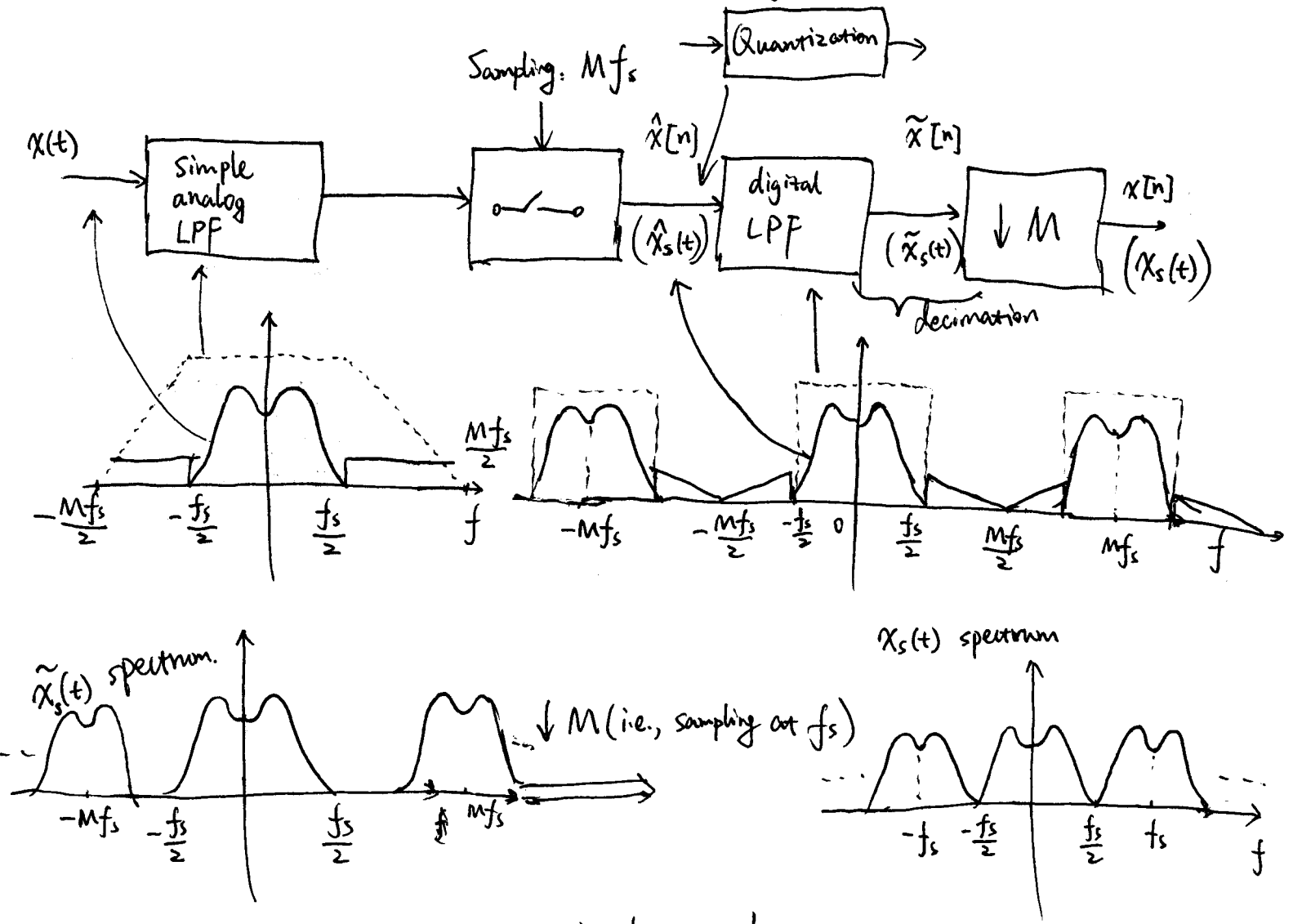


Problems: 1) Analog LPF with sharp cut-off frequency is difficult to implement. Big and expensive!

2) Multi-bit Quantizer is difficult to build.

Solution: Oversampling!

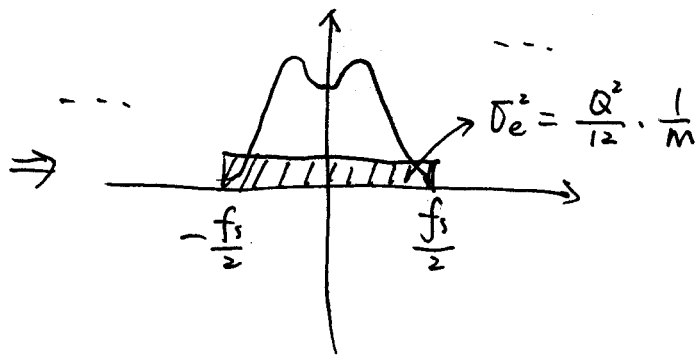
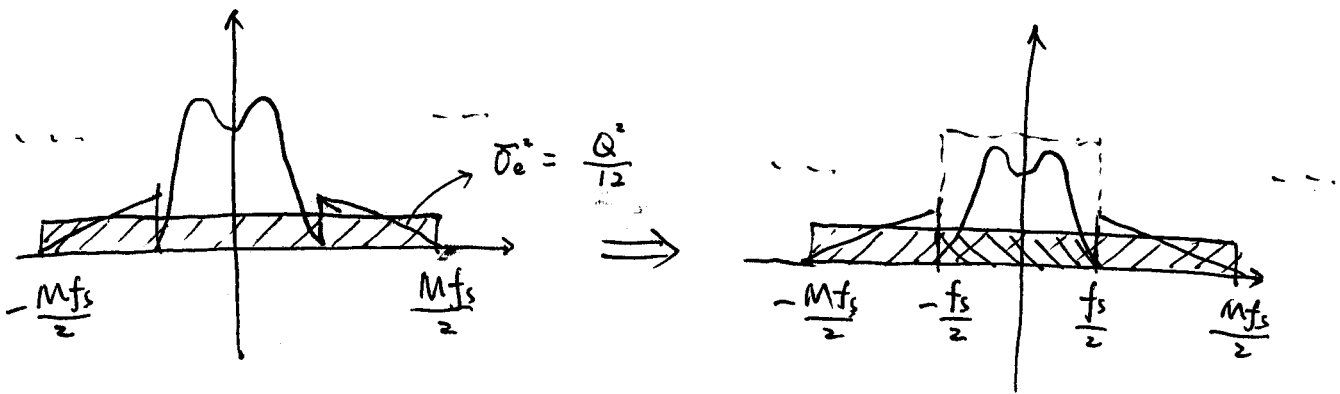
Suppose we want to retain the frequency components between 0 and  $\frac{f_s}{2}$  Hz of the input  $X(t)$ , we can do the following:



Digital LPF is much easier to implement!

### Oversampling increases quantization SNR

- 1) Assuming quantization noise has uniform distribution, <sup>PDF</sup> we have seen its power is  $\frac{Q^2}{12}$ .
- 2) Assuming PSD of quantization error is flat, i.e., uniformly distributed over entire frequency range, then power density is  $\frac{Q^2}{12} \cdot \frac{1}{Mf_s}$ .
- 3) After digital LPF with cut-off frequency of  $\frac{f_s}{2}$ , only part of the quantization noise is remained, while the entire signal energy is remain.



$\therefore$  SNR increase =  $10 \log_{10} M$

Oversampling factor	SNR increase
$M$	

$2$	$3 \text{ dB}$	
$2^2 = 4$	$6 \text{ dB}$	$\leftarrow$ Quantization level increase 1 bit
$2^3$	$9 \text{ dB}$	
$2^4 = 4^2$	$12 \text{ dB}$	$\leftarrow$ 2 bits

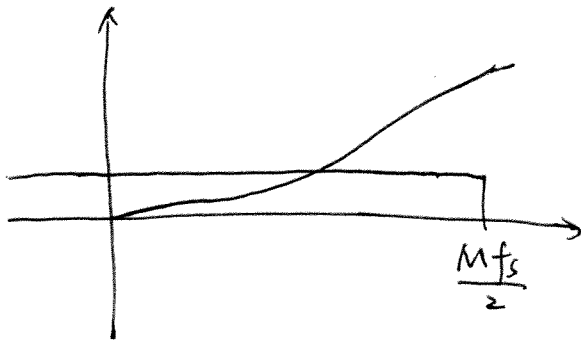
$\therefore$  Oversampling for  $4^N$  times is equivalent to increasing quantization bit depth for  $N$  bits.

Not very efficient. Can we do better?

Idea: Shape the quantization noise spectral shape!

(8)

If we can emphasize high-frequency part of noise, ~~it'd~~ and not change the signal.



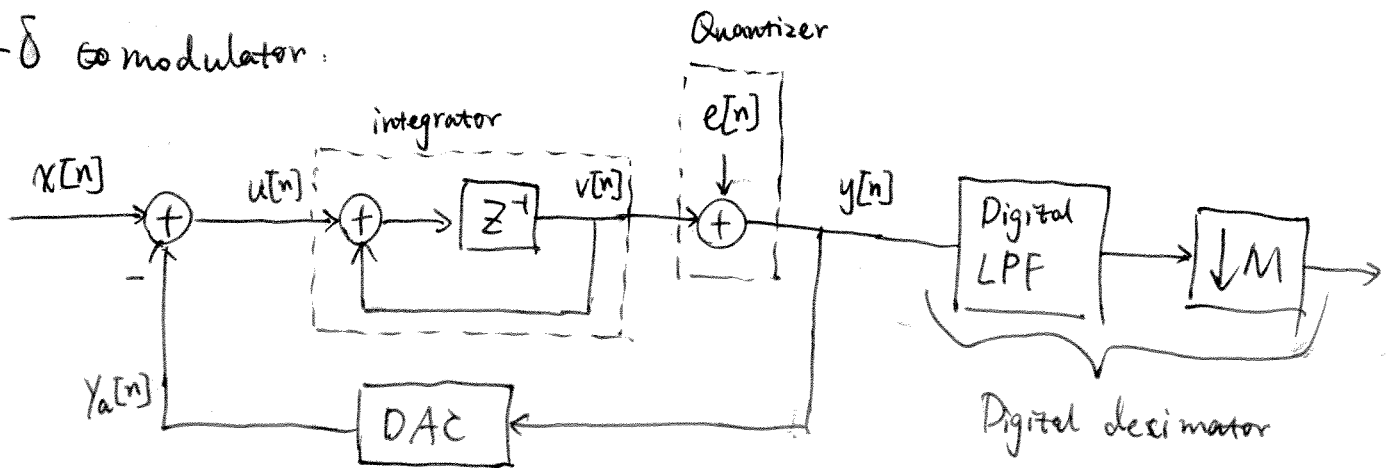
Quantization noise:  $X_Q[n] = X[n] + e[n]$

Z-transform:  $Y(z) = X(z) + E(z)$

In general:  $Y(z) = H_X(z) \cdot X(z) + H_E(z) E(z)$   
 ↑ transfer function for signal.      ↑ transfer function for noise

If we can design  $H_X(z)$  as all-pass filter and  $H_E(z)$  as high-pass filter, it'd be great!

$\Sigma$ - $\Delta$  modulator:



Integrator:  $(U(z) + V(z))z^{-1} = V(z)$ ,  $\therefore \frac{V(z)}{U(z)} = \frac{z^{-1}}{1-z^{-1}}$

Ideal DAC:  $Y_a[n] = y[n]$

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

⑨

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

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

$$\therefore H_X(z) = z^{-1} \quad \text{Delay one sample. (All-pass).}$$

$$H_E(z) = 1 - z^{-1} \quad \text{First-order differentiator (High-pass).}$$

Time domain  $y[n] = x[n-1] + e[n] - e[n-1]$

Power Density Spectrum of ~~y[n]~~ is shaped quantization noise  $e[n] = e[n] - e[n-1]$

is  $P_{E_i}(f) = P_E(f) \cdot |H_E(f)|^2$

or  $P_{E_i}(e^{j\Omega}) = P_E(e^{j\Omega}) \cdot |H_E(e^{j\Omega})|^2 = P_E(e^{j\Omega}) / |1 - e^{j\Omega}|^2$

$$= P_E(e^{j\Omega}) \left| e^{j\frac{\Omega}{2}} (e^{-j\frac{\Omega}{2}} - e^{j\frac{\Omega}{2}}) \right|^2$$

$$= P_E(e^{j\Omega}) \cdot 4 \sin^2 \frac{\Omega}{2}$$

After Digital LPF with cutoff frequency of  $\frac{f_s}{2}$ , the rest noise power is

$$\sigma_{e_i}^2 = \int_{-\frac{f_s}{2}}^{\frac{f_s}{2}} P_{E_i}(f) df = P_E(f) \int_{-\frac{f_s}{2}}^{\frac{f_s}{2}} 4 \sin^2 \left( \frac{2\pi f / f_s \cdot M}{2} \right) df$$

$$= P_E(f) \cdot 2 \int_0^{f_s/2} 4 \sin^2 \left( \pi \frac{f}{M f_s} \right) df$$

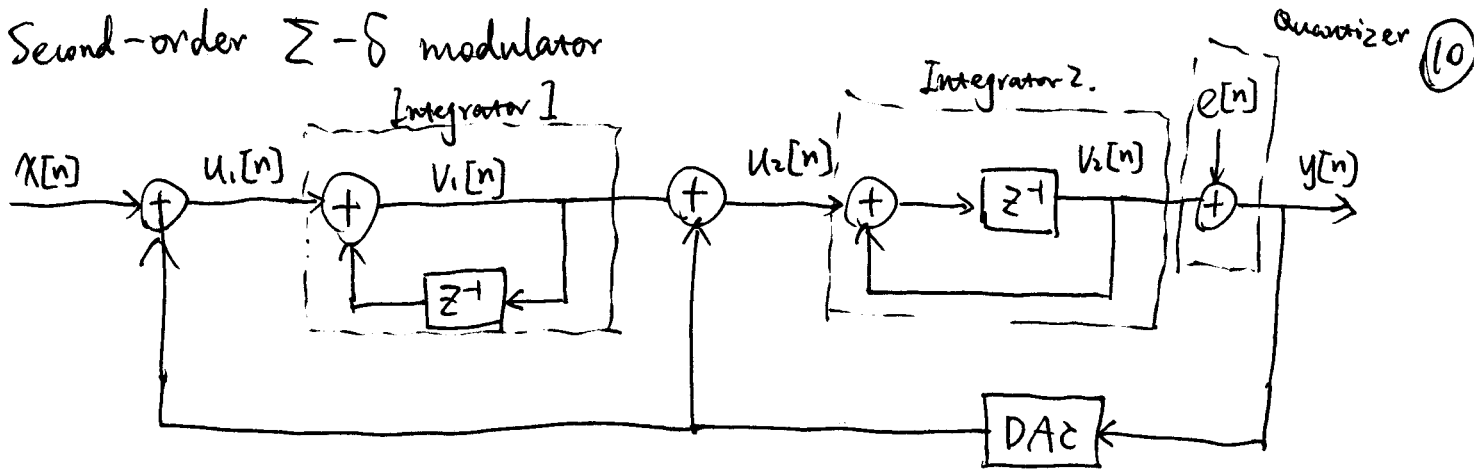
$$= \frac{Q^2}{12} \cdot \frac{\pi^2}{3} \cdot \frac{1}{M^3}$$

$$\therefore \text{SNR increase} : 10 \log_{10}(M^3) - 10 \log_{10} \left( \frac{\pi^2}{3} \right)$$

$$= 30 \log_{10} M - 10 \log_{10} \left( \frac{\pi^2}{3} \right)$$

M	SNR increase	equivalent bit depth increase
2	9dB	1.5 bits
4	18dB	3 bits
⋮	⋮	⋮

# Second-order $\Sigma$ - $\Delta$ modulator



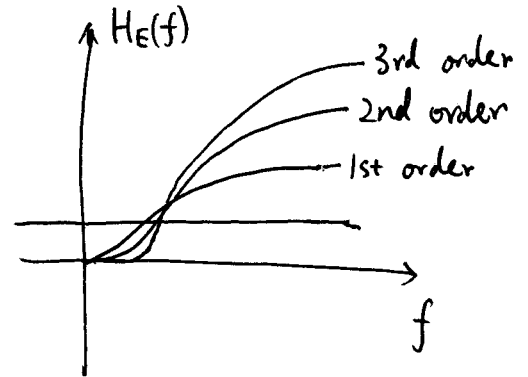
Integrator 1:  $\frac{V_1(z)}{U_1(z)} = \frac{1}{1-z^{-1}}$

Integrator 2:  $\frac{V_2(z)}{U_2(z)} = \frac{z^{-1}}{1-z^{-1}}$

$$Y(z) = X(z)z^{-1} + E(z)(1-z^{-1})^2$$

SNR increase:  $50 \log_{10} M - 10 \log_{10} \left( \frac{\pi^4}{5} \right)$

$\therefore$  Doubling oversampling factor gives  $\sim 15$  dB SNR increase, i.e., about 2.5 bits quantization bit-depth increase.



$L$ -order  $\Sigma$ - $\Delta$  modulator: Have  $L-1$  Integrators of  $\frac{1}{1-z^{-1}}$ , then  
1 integrator of  $\frac{z^{-1}}{1-z^{-1}}$

Noise transfer function  $H_E(z) = (1-z^{-1})^L$

signal - - - - -  $H_X(z) = z^{-1}$

SNR increase  $\approx \frac{2L+1}{L} \approx \underline{\underline{(2L+1) 10 \log_{10} M - 10 \log_{10} \left( \frac{\pi^2 \pi^{2L}}{2L+1} \right)}}$

Doubling oversample factor  $\sim$  SNR increase  $(6L+3)$  dB  $\sim$  bit depth increase  $\frac{6L+3}{2} \approx (L+0.5)$  bits