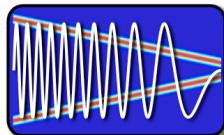


EE123

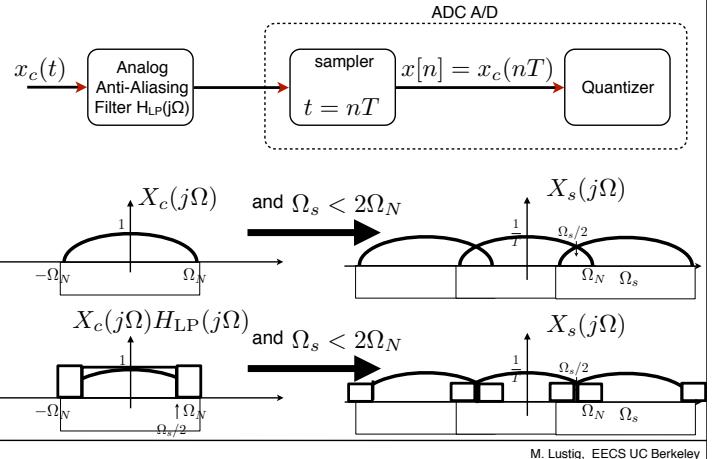


# Digital Signal Processing

## Lecture 19 Practical ADC/DAC

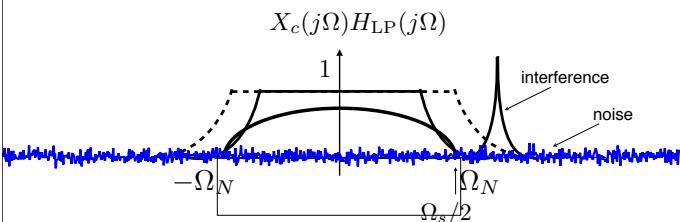
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### Ideal Anti-Aliasing



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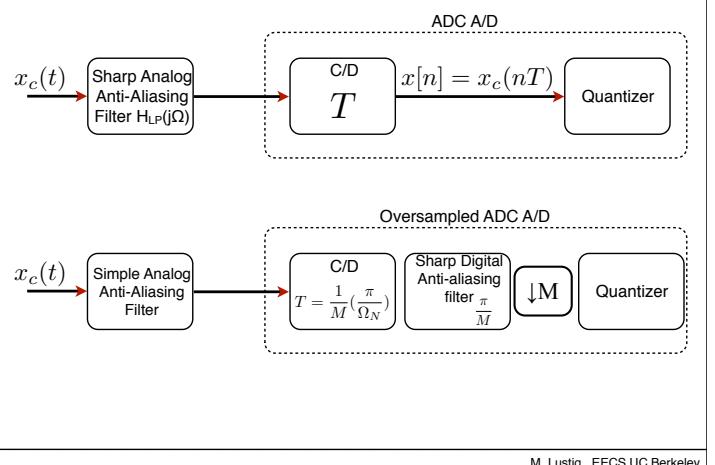
### Non Ideal Anti-Aliasing



- Problem: Hard to implement sharp analog filter
- Tradeoff:
  - Crop part of the signal
  - Suffer from noise and interference (See lab II !)

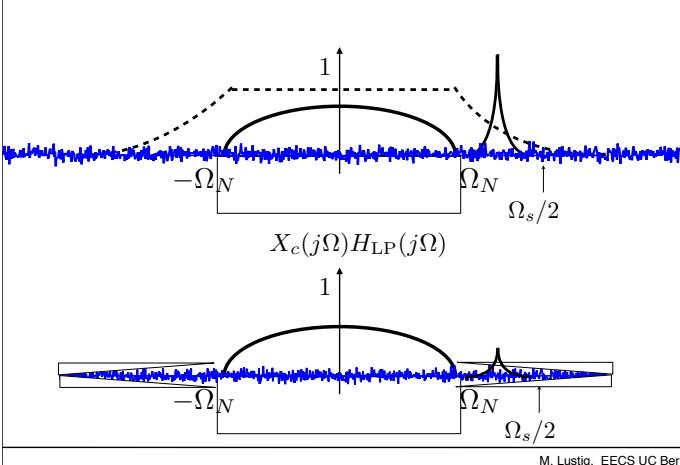
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### Oversampled ADC



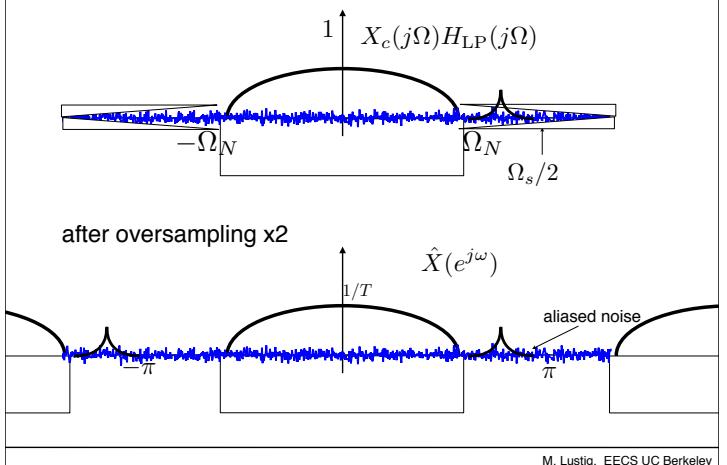
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### Oversampled ADC



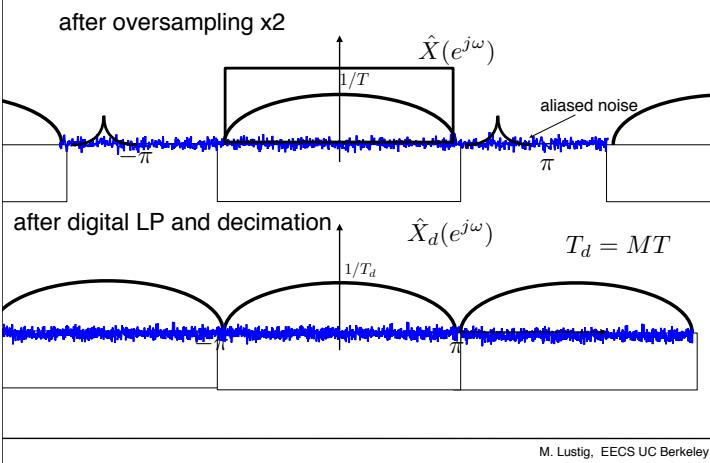
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### Oversampled ADC

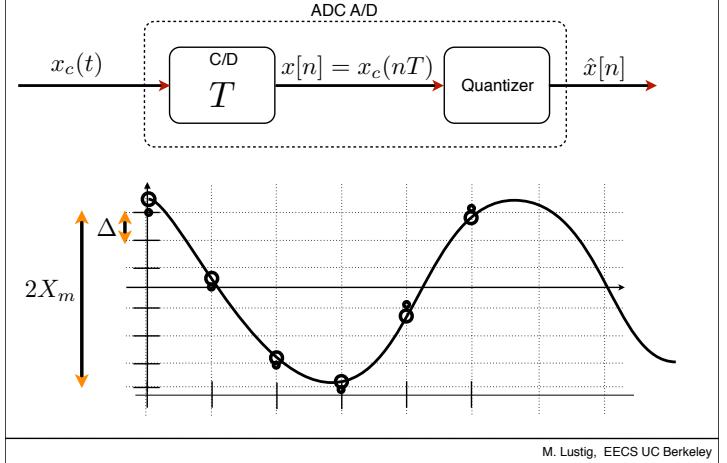


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## Oversampled ADC



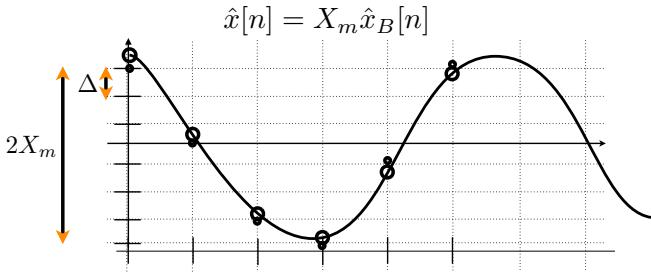
## Sampling and Quantization



## Sampling and Quantization

- for 2's complement with  $B+1$  bits  $-1 \leq \hat{x}_B[n] < 1$

$$\Delta = \frac{2X_m}{2^{B+1}} = \frac{X_m}{2^B}$$



## Quantization Error



- Model quantization error as noise

$$x[n] \xrightarrow{+} \hat{x}[n] = x[n] + e[n]$$

$e[n]$

- In that case:

$$-\Delta/2 \leq e[n] < \Delta/2$$

$$(-X_m - \Delta/2) < x[n] \leq (X_m - \Delta/2)$$

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## Noise Model for Quantization Error

### Assumptions:

- Model  $e[n]$  as a sample sequence of a stationary random process
- $e[n]$  is not correlated with  $x[n]$ , e.g.,  $E[e[n]x[n]] = 0$
- $e[n]$  not correlated with  $e[m]$ , e.g.,  $E[e[n]e[m]] = 0 \mid m \neq n$  (white noise)
- $e[n] \sim U[-\Delta/2, \Delta/2]$

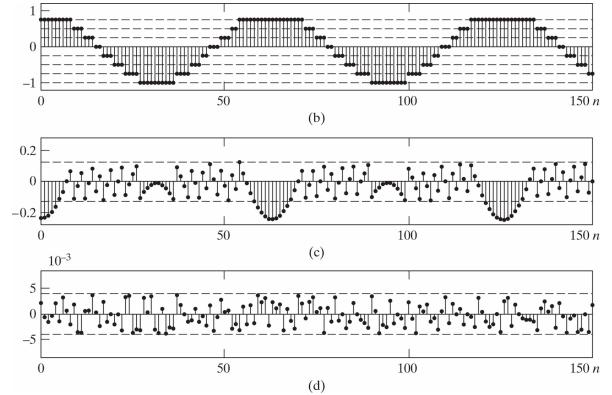
### Result:

- Variance is:  $\sigma_e^2 = \frac{\Delta^2}{12}$ , or  $\sigma_e^2 = \frac{2^{-2B}X_m^2}{12}$  since  $\Delta = 2^{-B}X_m$
- Assumptions work well for signals that change rapidly, are not clipped and for small  $\Delta$

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## Quantization Noise

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).



## SNR of Quantization Noise

- For uniform B+1 bits quantizer:  $\sigma_e^2 = \frac{2^{-2B} X_m^2}{12}$

$$\begin{aligned} SNR_Q &= 10 \log_{10} \left( \frac{\sigma_x^2}{\sigma_e^2} \right) \\ &= 10 \log_{10} \left( \frac{12 \cdot 2^{2B} \sigma_x^2}{X_m^2} \right) \end{aligned}$$

$$SNR_Q = 6.02B + 10.8 - 20 \log_{10} \left( \frac{X_m}{\sigma_x} \right) \text{ Quantizer range rms of amp}$$

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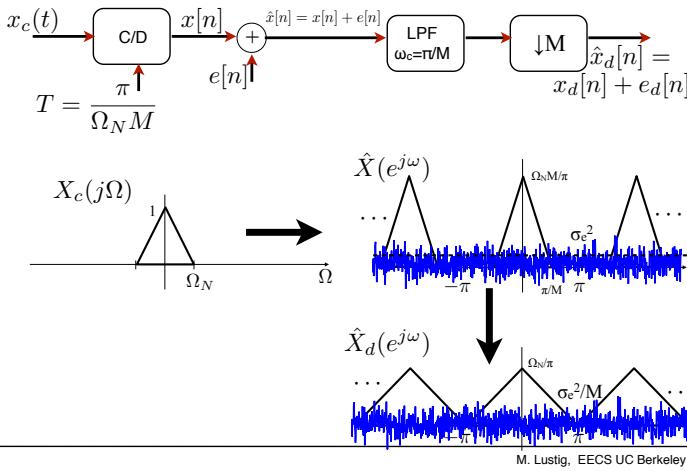
## SNR of Quantization Noise

$$SNR_Q = 6.02B + 10.8 - 20 \log_{10} \left( \frac{X_m}{\sigma_x} \right) \text{ Quantizer range rms of amp}$$

- Improvement of 6dB with every bit
- The range of the quantization must be adapted to the rms amplitude of the signal
  - Tradeoff between clipping and noise!
  - Often use pre-amp
  - Sometimes use analog auto gain controller (AGC)
  - If  $\sigma_x = X_m/4$  then  $SNR_Q \approx 6B - 1.25dB$   
so SNR of 90-96 dB requires 16-bits (audio)

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## Quantization noise in Oversampled ADC



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## Quantization noise in Oversampled ADC

- Energy of  $x_d[n]$  equals energy of  $x[n]$ 
  - No filtering of signal!
- Noise std is reduced by factor of M

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

- For doubling of M we get 3dB improvement, which is the same as 1/2 a bit of accuracy
  - With oversampling of 16 with 8bit ADC we get the same quantization noise as 10bit ADC!

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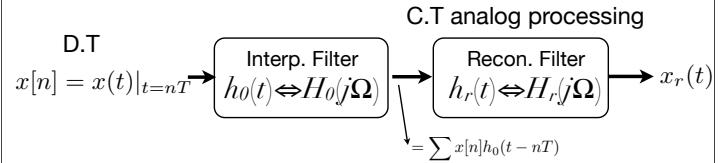
## Practical ADC (Ch. 4.8.4)

$$\text{D.T } x[n] = x(t)|_{t=nT} \rightarrow \text{sinc pulse generator} \rightarrow x_r(t) = \sum_{n=-\infty}^{\infty} x[n] \text{sinc} \left( \frac{t-nT}{T} \right)$$

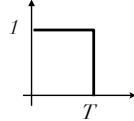
- Scaled train of sinc pulses
- Difficult to generate sinc  $\Rightarrow$  Too long!

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## Practical ADC



- $h_0(t)$  is finite length pulse  $\Rightarrow$  easy to implement
- For example: zero-order hold

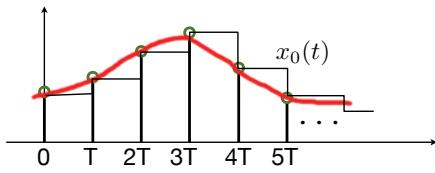


$$H_0(j\Omega) = Te^{-j\Omega \frac{T}{2}} \text{sinc} \left( \frac{\Omega}{\Omega_s} \right)$$

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## Practical ADC

Zero-Order-Hold interpolation



$$x_0(t) = \sum_{n=-\infty}^{\infty} x[n]h_0(t-nT) = x_0(t) * x_s(t)$$

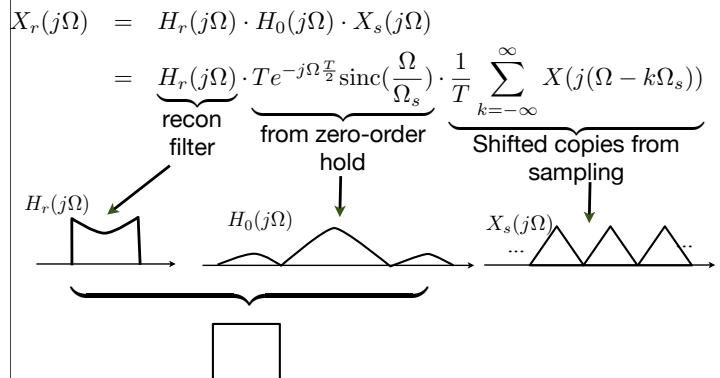
Taking a FT:

$$\begin{aligned} X(j\Omega) &= H_0(j\Omega)X_s(j\Omega) \\ &= H_0(j\Omega)\frac{1}{T}\sum_{k=-\infty}^{\infty} X(j(\Omega - k\Omega_s)) \end{aligned}$$

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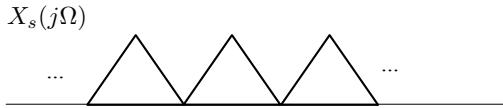
## Practical ADC

Output of the reconstruction filter:

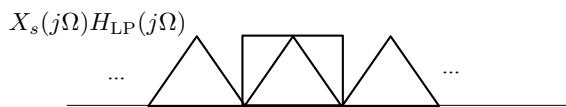


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## Practical ADC

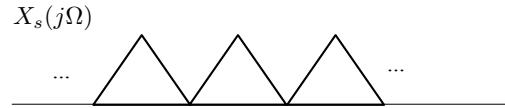


Ideally:

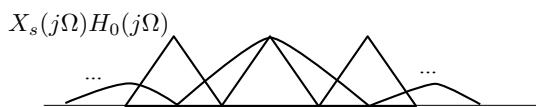


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## Practical ADC

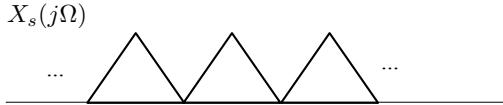


Practically:

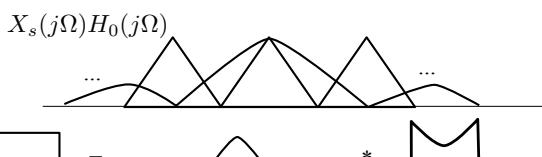


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## Practical ADC

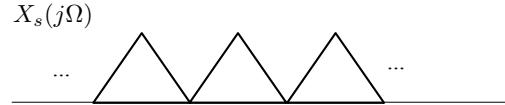


Practically:

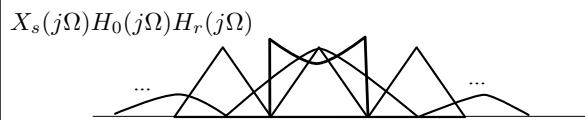


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## Practical ADC

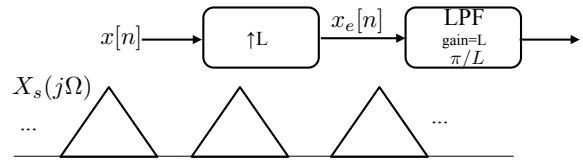


Practically:

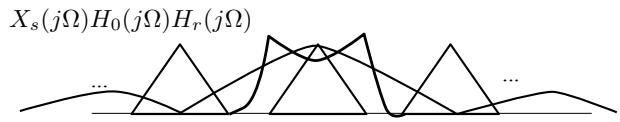


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### Easier Implementation with Digital upsampling



Practically:



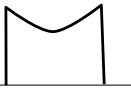
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### Easier Implementation with Digital upsampling

easier implementing  
with analog components



Need analog components  
made of Nonobtainium



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