- Impulse invariance
- Sampling of the impulse response of LTI CT systems

$$h[n] = Th_c(nT),$$
 $H(e^{j\omega}) = H_c\left(j\frac{\omega}{T}\right), \qquad |\omega| \le \pi.$

Example: LPF

$$H_{\mathcal{C}}(j\Omega) = \begin{cases} 1, & |\Omega| < \Omega_{\mathcal{C}}, \\ 0, & |\Omega| \leq \Omega_{\mathcal{C}}. \end{cases}$$

$$\Omega_{c} = \omega_{c}/T < \pi/T$$

The impulse response of the CT LPF:

$$h_c(t) = \frac{\sin(\Omega_c t)}{\pi t}$$

The impulse response of DT system

$$h[n] = Th_c(nT) = T\frac{\sin(\Omega_c nT)}{\pi nT} = \frac{\sin(\omega_c n)}{\pi n}$$

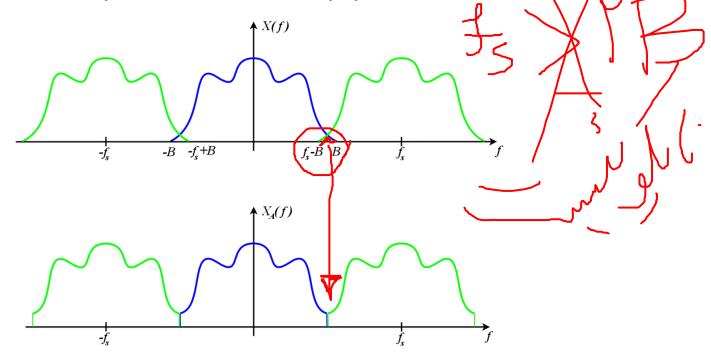
And,

$$\omega_c = \Omega_c T$$

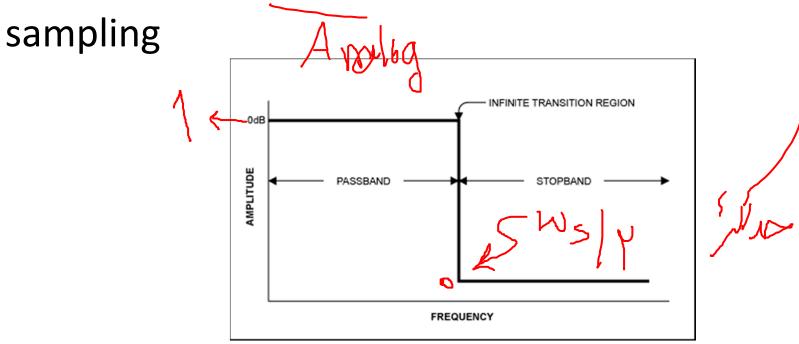
$$H(e^{j\omega}) = \begin{cases} 1, & |\omega| < \omega_{c}, \\ 0, & \omega_{c} < |\omega| \le \pi. \end{cases}$$

Anti-aliasing filter

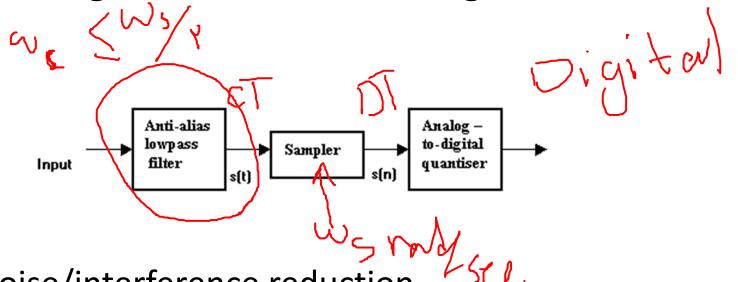
Aliased components and Nyquest theorem



 Given a sampling frequency: properly selection of the LPF to ensure "secure"



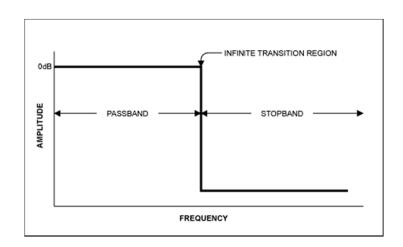
Block diagram with anti-aliasing filter

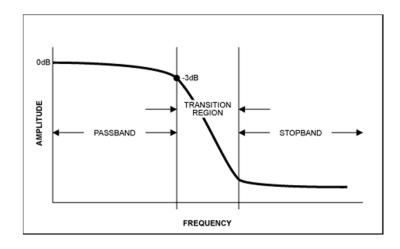


Noise/interference reduction

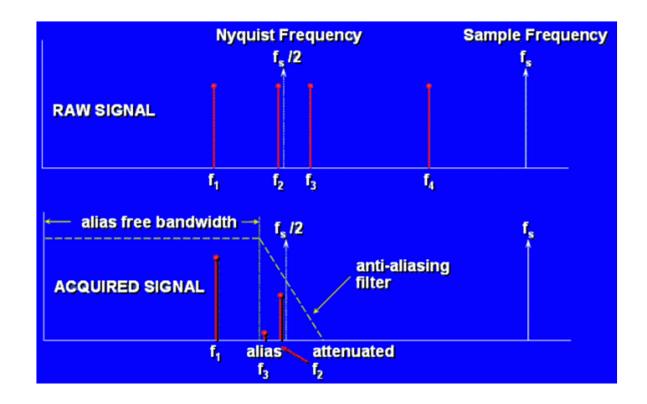
https://www2.spsc.tugraz.at/www-archive/AdvancedSignalProcessing/SS01-VoiceOverIP/coders/quantization.html

- Ideal vs. non-ideal filters
 - Ideal filters are not realizable

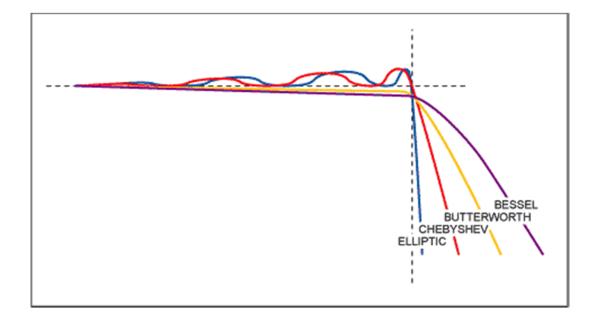




An example of the non-ideal LPF



- Addressing filter design
 - Various methods
 - Passband
 - Stopband



- Changing the sampling rate
 - Sampling of continuous-time signals

$$x[n] = x_c(nT)$$

Reconstruction: interpolation formula

$$x_r(t) = \sum_{n=-\infty}^{\infty} x[n]h_r(t-nT).$$

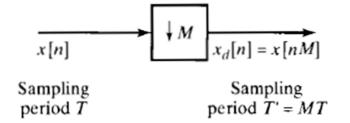
$$x_r(t) = \sum_{n=-\infty}^{\infty} x[n] \frac{\sin[\pi(t-nT)/T]}{\pi(t-nT)/T}$$

Changing the sampling rate

$$x[n] = x_c(nT) \qquad x'[n] = x_c(nT')$$

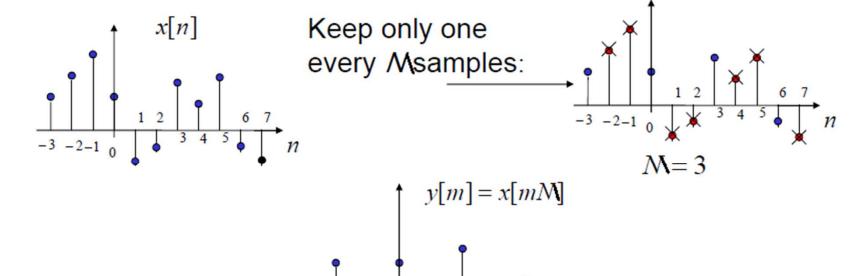
- A suggestion:
 - 1) Reconstruction of the CT signal
 - 2) sampling with desired rate
 - Associated with non-idealities
 - Complexity: ADC, DAC

- We consider changing the sampling rate in discrete-time domain
- Sampling rate reduction by an integer factor
 - Downsampling/decimation



$$x_d[n] = x[nM] = x_c(nMT)$$

- Compressor system: integer M
 - Discarding M-1 samples out of M



The compressor is not TI

$$y[n] = x[Mn], \quad -\infty < n < \infty,$$

— The response to $x_1[n] = x[n - n_0]$

$$y_1[n] = x_1[Mn] = x[Mn - n_0].$$

Compared with

$$y[n-n_0] = x[M(n-n_0)].$$

Investigation

- Let $X_c(j\Omega) = 0$ for $|\Omega| \ge \Omega_N$
- When can we reduce the sampling rate by factor M?
- The original sampling frequency must be at least
 M times of the Nyquest rate
- More on this later!

- Frequency analysis
 - We had $x[n] = x_c(nT)$

$$X(e^{j\omega}) = \frac{1}{T} \sum_{k=-\infty}^{\infty} X_{c} \left(j \left(\frac{\omega}{T} - \frac{2\pi k}{T} \right) \right)$$

— Therefore, for $x_d[n] = x[nM] = x_c(nT')$

$$X_d(e^{j\omega}) = \frac{1}{T'} \sum_{r=-\infty}^{\infty} X_c \left(j \left(\frac{\omega}{T'} - \frac{2\pi r}{T'} \right) \right)$$

interpretation

• Next consider that T' = MT

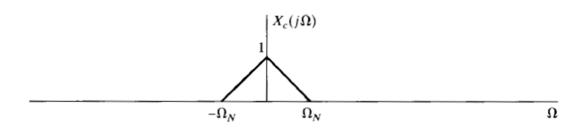
$$X_d(e^{j\omega}) = \frac{1}{MT} \sum_{r=-\infty}^{\infty} X_c \left(j \left(\frac{\omega}{MT} - \frac{2\pi r}{MT} \right) \right)$$

• Finally, it can be shown that

$$X_d(e^{j\omega}) = \frac{1}{M} \sum_{i=0}^{M-1} X(e^{j(\omega/M - 2\pi i/M)})$$

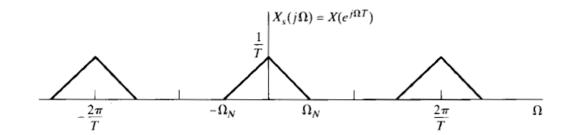
– M copies!

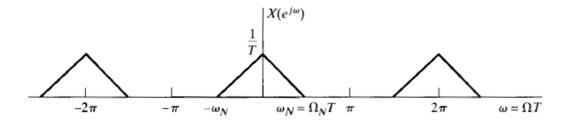
Illustration



$$2\pi/T = 4\Omega_N$$

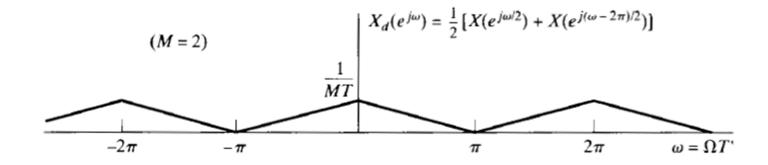
$$\omega_N = \Omega_N T = \pi/2$$



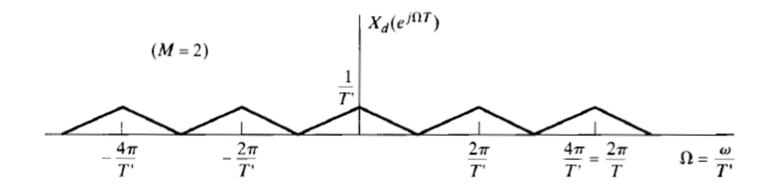


And after, downsampling

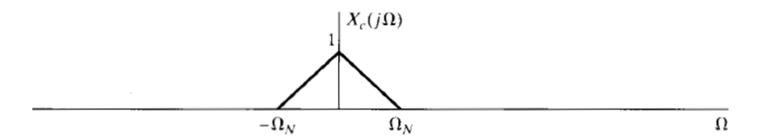
$$2\pi/T = 4\Omega_N$$



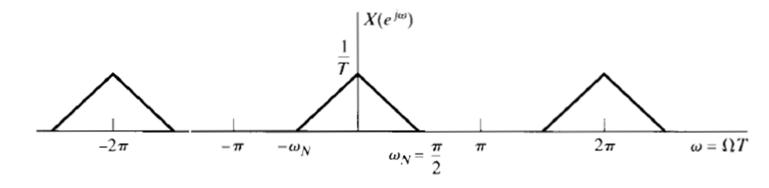
$$X_d(e^{j\omega}) = \frac{1}{M} \sum_{i=0}^{M-1} X(e^{j(\omega/M - 2\pi i/M)})$$

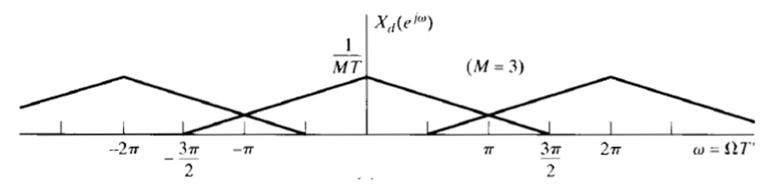


An example of aliasing via downsampling



• Cont.





Consequently, for safe downsampling we should have

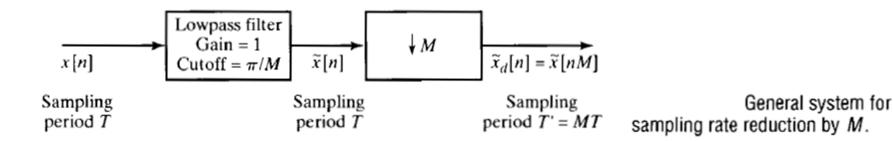
$$X(e^{j\omega}) = 0, \qquad \omega_N \le |\omega| \le \pi.$$

And

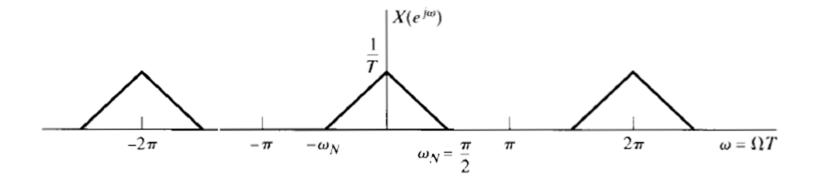
$$\omega_N < \pi/M$$

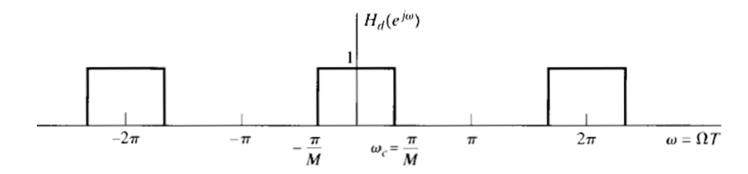
- Check it!
- Otherwise, we will have aliasing!!

- Avoiding aliasing in downsamplin via DT antialiasing filter
- The cut-off frequency: $\omega_N < \pi/M$



Modified version of the aliased example





• Cont.

