Introduction to Digital Signal Processing Sampling theorem revisited

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Sampling theorem

$$x(t) \longrightarrow x[n] \longrightarrow x(t)$$

This is possible if the **signal is bandlimited** \implies limit on the how fast the signal varies.

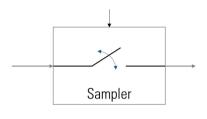
A measure of how fast the signal varies \longrightarrow Max. frequency component.

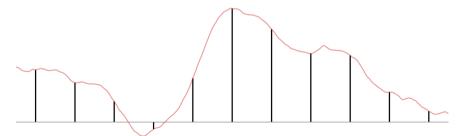
Nyquist-Shanon Sampling Theorem.

If a signal x(t) contains no frequencies higher than f_{sig} Hz, then it is completely determined by its values at time points spaced less than $1/(2f_{sig})$ seconds apart.

$$\implies$$
 Sampling rate $=F_s>2f_{sig}$

Sampling process





Fourier transform of an Impulse train

$$\sum_{n=-\infty}^{\infty} \delta\left(t - n \cdot T_s\right) \stackrel{\mathsf{FT}}{\longleftrightarrow} \frac{2\pi}{T_s} \sum_{n=-\infty}^{\infty} \delta\left(\omega - n \cdot \omega_s\right)$$

Let's say we have a discrete-time signal x[n] that was obtained from sampling a continuous-time signal x(t), such that

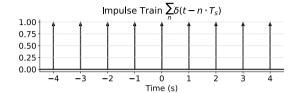
$$x[n] = x(n \cdot T_s), \forall n \in \mathbb{Z}$$

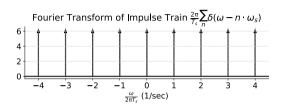
One way to reconstruct x(t) from x[n] is to generate an impulse train first,

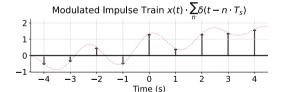
$$x_{\delta}(t) = \sum_{n} x[n] \cdot \delta(t - n \cdot T_s) = x(t) \cdot \sum_{n} \delta(t - n \cdot T_s)$$

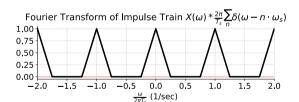
$$x(t) \cdot \sum_{n} \delta(t - n \cdot T_s) \stackrel{\mathsf{FT}}{\longleftrightarrow} X(\omega) * \frac{2\pi}{T_s} \sum_{n = -\infty}^{\infty} \delta(\omega - n \cdot \omega_s)$$

Fourier transform of a Modulated Impulse train

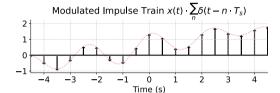


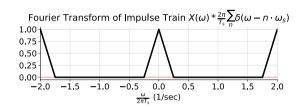




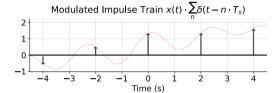


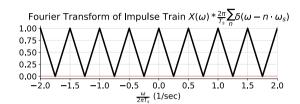
Fourier transform of a Modulated Impulse train





Sampling at the Nyquist rate





What happens when the signal is not bandlimited?

