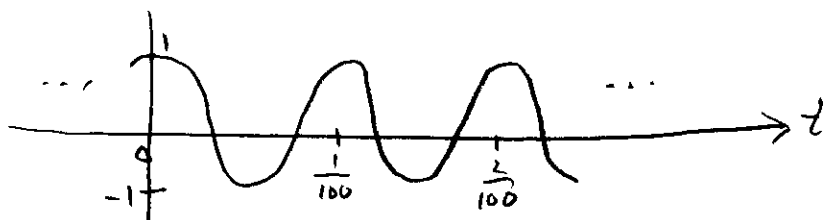


Getting Started

Sampled (Discrete-Time) Signals

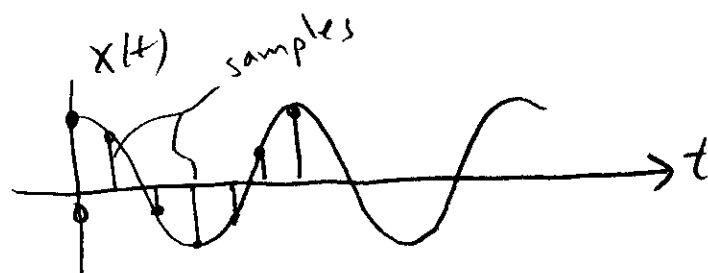
You are used to seeing and working with continuous-time (CT) signals. For example,

$$x(t) = \cos(2\pi(100)t)$$



In Digital Signal Processing we focus on discrete-time (DT) signals.

$$x(t) = \cos(2\pi(100)t)$$



$x[n]$

n = sample index, acts like time
 n = integer

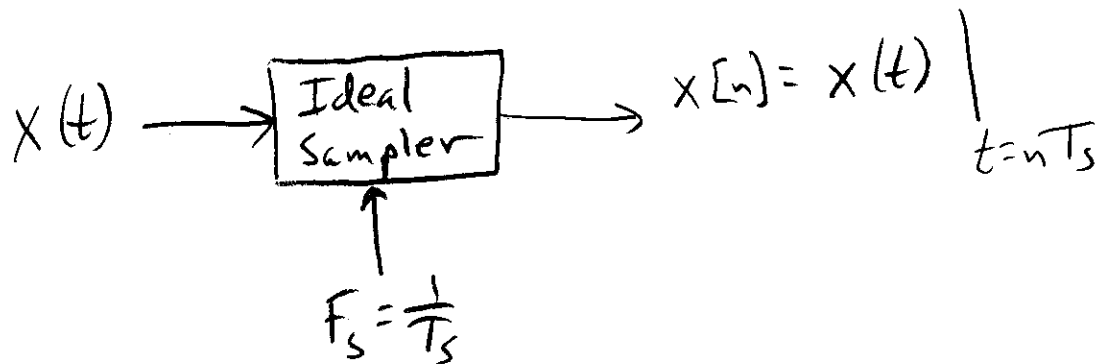


Q: Why do we focus on DT signals?

A: We can't use computers to process C.T. signals, but we can use them for DT signals.

Q: How do we get a D.T. signal from a C.T. signal?

A: Sampling



Ideal means the samples are infinite precision. That is to say, the number of bits/sample, B , is infinity.

Notice that in sampling, we lose information.
We don't keep what happens between the samples.

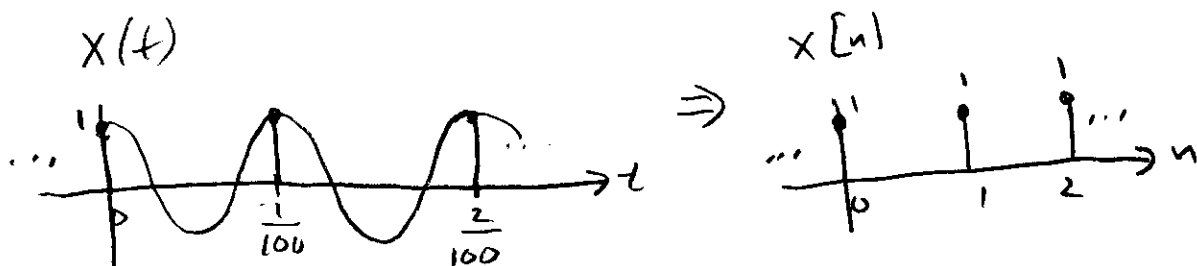
What are some consequences of this loss of information?

Consider an example.

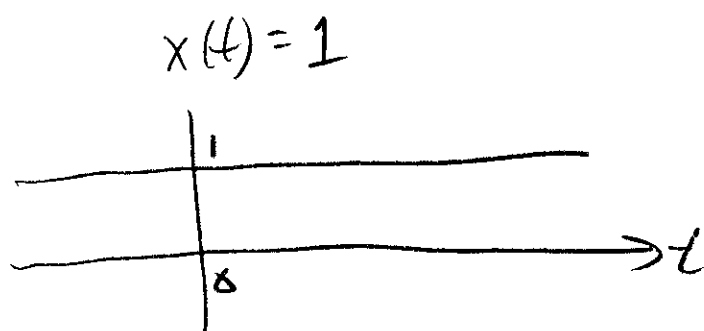
$$x(t) = \cos(2\pi(100)t) \quad , \quad \text{period} = T_s = \frac{1}{100} \text{ sec.}$$

Now sample at exactly this period.

$$\begin{aligned} x[n] &= x(t) \Big|_{t=nT_s = \frac{n}{100}} \\ &= \cos\left(2\pi(100)\frac{n}{100}\right) \\ &= \cos(2\pi n) = 1 \quad \text{for all } n. \end{aligned}$$



Notice $x[n]$ is exactly the same as if the original $x(t)$ had been the constant 1 for all time.



Thus, we can't tell $\cos(2\pi(100)t)$ from 1 if $T_s = \frac{1}{100}$.
(aliasing)

It turns out there is a limit on the range of frequencies that we can distinguish given $F_s = \frac{1}{T_s}$.

Shannon's Sample Theorem (Claude Shannon)

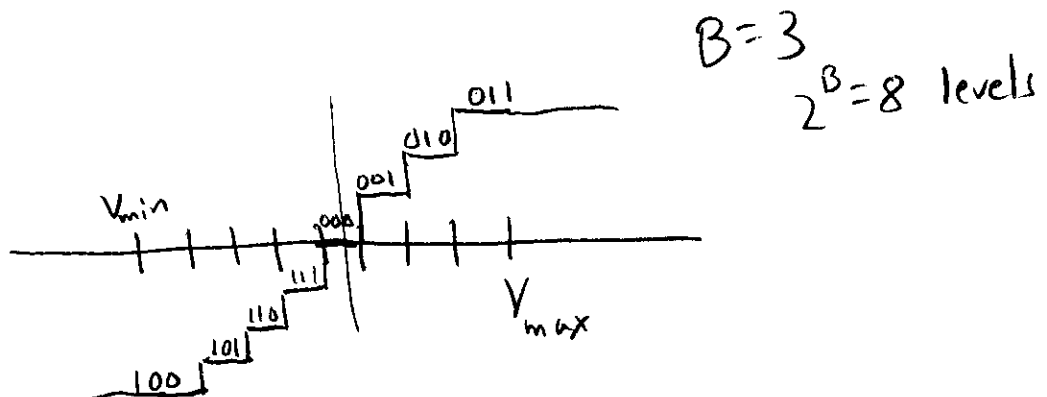
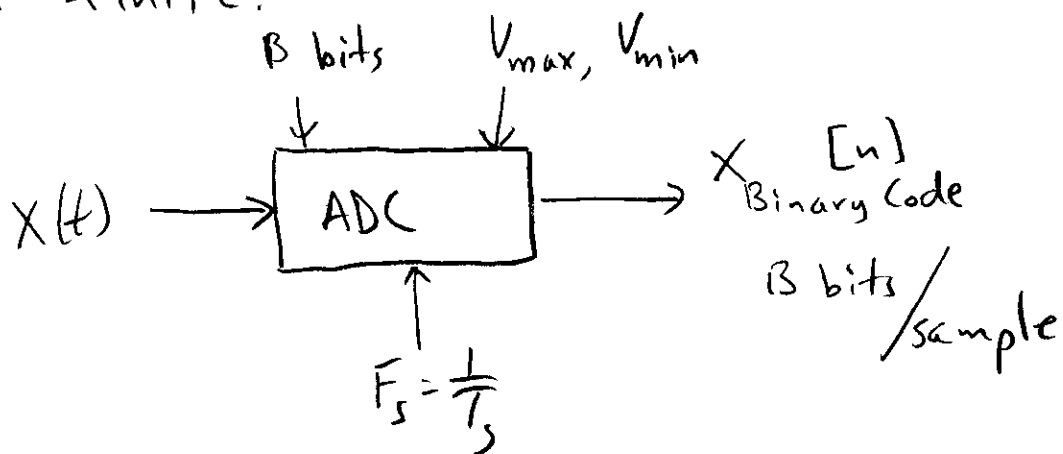
If $x(t)$ is bandlimited, such that the highest frequency in $x(t)$ is F_{\max} , then $x(t)$ can be exactly reconstructed from its samples if

$$F_s = \frac{1}{T_s} > 2F_{\max}$$

$2F_{\max}$ = Nyquist Frequency

Ideal samplers (i.e., infinite precision samplers) do not exist in real life.

However Analog to Digital Converters (ADCs) do exist. For the the number of bits/sample is finite.



In this case the output is a 2's complement signed binary integer.

This is not necessarily the same number as the value of $X(t)$, but a re-mapping onto integers.

ADC examples you may have seen.

PC sound chips (audio codec = coder/decoder)

2 channels \rightarrow stereo

F_s typically up to 44100 Hz, 48 kHz, 96 kHz

$B = 16, 20, 24$ or more

Digital Audio Standard for CDs
16 bits, $F_s = 44100$ Hz

HDCD
24 bits, $F_s = 44100$ Hz

Some DVD audio (5.1 surround sound)
24 bits $F_s = 48$ kHz or 96 kHz

24 \rightarrow audiophile quality

Raspberry Pi - audiophile DAC Hats

igaudio \rightarrow Geekworm kit 24 bits } Texas Instruments DAC
inno-make \rightarrow 32 bits

Hifiberry

DAC+ ADC pro

DAC2 pro

DAC2 HD

} 24 bits
Burr Brown DAC

The DAC2 Hats can also use a
DSP chip add-on board for real-time
filtering