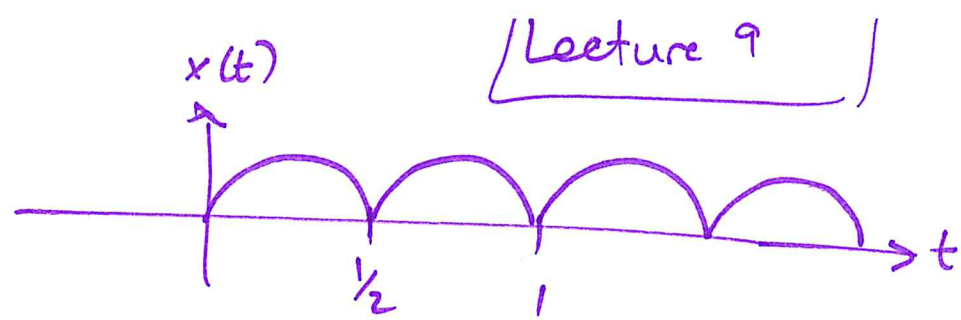


Example from last time:

$$x(t) = |\sin(2\pi t)|$$

$$T_0 = \frac{1}{2} \text{ s.}$$

$$f_0 = \frac{1}{T_0} = 2 \text{ Hz.}$$



①

$$a_k = \frac{1}{T_0} \int_0^{T_0} x(t) e^{-j2\pi k f_0 t} dt$$

$$= \int_0^{1/2} \left[\frac{1}{j} e^{j2\pi t(1-2k)} \rightleftharpoons \frac{1}{j} e^{-j2\pi t(1+2k)} \right] dt.$$

$$= \frac{1}{j} \frac{e^{j2\pi t(1-2k)}}{j2\pi(1-2k)} \Big|_0^{1/2} + \frac{1}{j} \frac{e^{-j2\pi t(1+2k)}}{+j2\pi(1+2k)} \Big|_0^{1/2}$$

$$\textcircled{\Psi} = \left(\frac{-e^{j4\pi(1-2k)}}{2\pi(1-2k)} \right) - \left(\frac{-1}{2\pi(1-2k)} \right) + \left(\frac{-e^{-j\pi(1+2k)}}{2\pi(1+2k)} \right) - \left(\frac{-1}{2\pi(1+2k)} \right)$$

$$j^2 = -1$$

Recall: $e^{j\pi} = -1$ $e^{-j2\pi k} = 1$ $e^{-j\pi} = \frac{1}{e^{j\pi}} = \frac{1}{-1} = -1$ (2)

$$\textcircled{*} = \frac{-e^{j\pi} \cdot e^{-j\pi 2k}}{2\pi(1-2k)} + \frac{1}{2\pi(1-2k)} - \frac{e^{-j\pi} e^{-j\pi 2k}}{2\pi(1+2k)} + \frac{1}{2\pi(1+2k)}$$

$$= \frac{-(-1)(1)}{2\pi(1-2k)} + \frac{1}{2\pi(1-2k)} - \frac{(-1)(1)}{2\pi(1+2k)} + \frac{1}{2\pi(1+2k)}$$

$$a_k = \frac{1}{\pi(1-2k)} + \frac{1}{\pi(1+2k)}$$

$$e^{j\theta} = \cos(\theta) + j\sin(\theta)$$

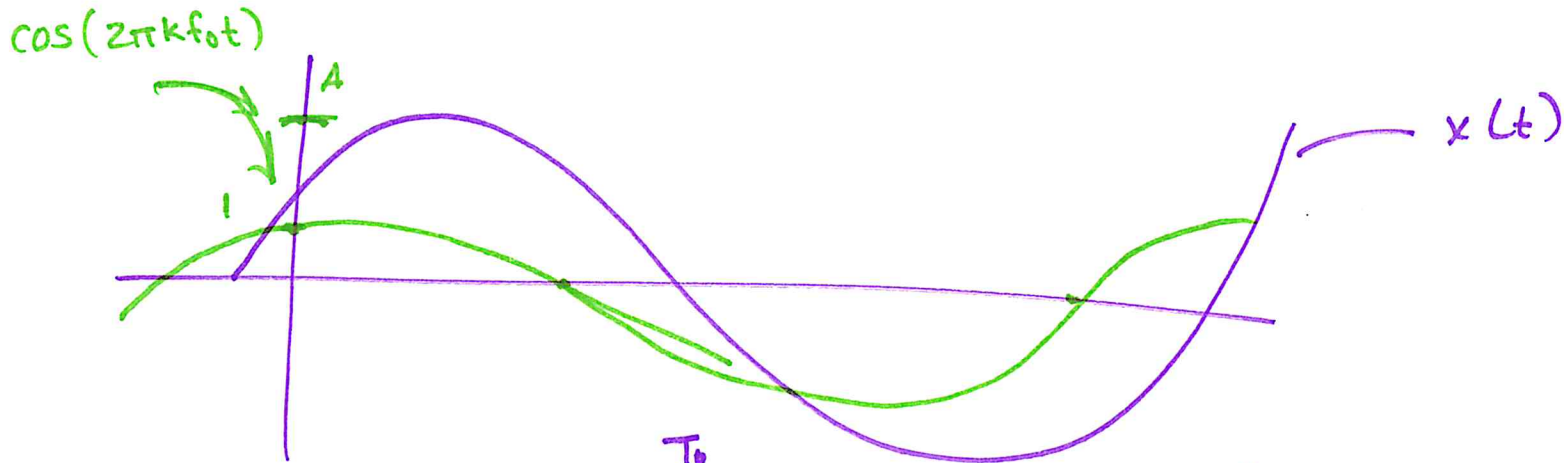
$$e^{j(-2\pi k)} = \cos(-2\pi k) + j\sin(-2\pi k) = 1$$

Note:

3

we set $\underline{a_k} = \frac{1}{T_0} \int_0^{T_0} x(t) e^{-j2\pi k f_0 t} dt$ $\textcircled{\star}$

the sinusoid $a_k \cdot e^{j2\pi k f_0 t}$ is the "best" fit to $x(t)$ @ freq $k f_0$



$$\text{let error}(a) = \int_0^{T_0} |x(t) - a e^{j2\pi k f_0 t}|^2 dt$$

we can show -that a_k from $\textcircled{\star}$ minimizes error(a)

$$\text{error}(a_k) \leq \text{error}(a) \text{ for all } a$$

Lecture 9

4

So far in class, continuous time / analog signals

signal $x(t)$ where t is real-valued

e.g. $x(t) = \cos(2\pi ft)$

In lab, we represent signals as list of numbers
for storing and processing

Ex 1 CD audio: 3 min song \approx 32 MB

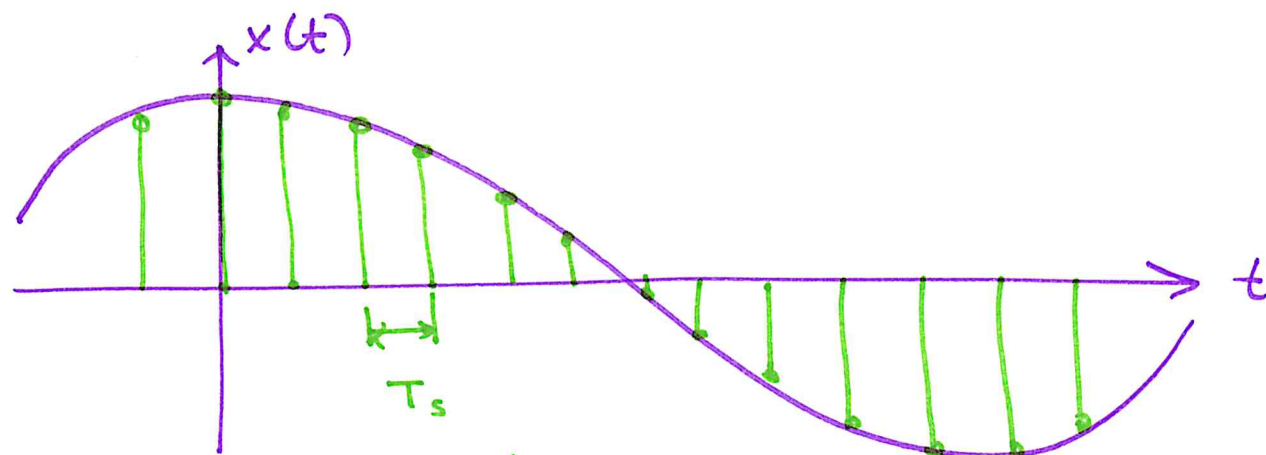
Ex 2 digital image — store individual pixels

Analog to Digital Conversion

convert a continuous-time signal to a digital or discrete-time signal

AKA "sampling"

Basic idea:



= sampling period

$\frac{1}{T_s} = f_s =$ sampling frequency.

$$x[n] = x(n \cdot T_s)$$



n^{th} discrete-time sample

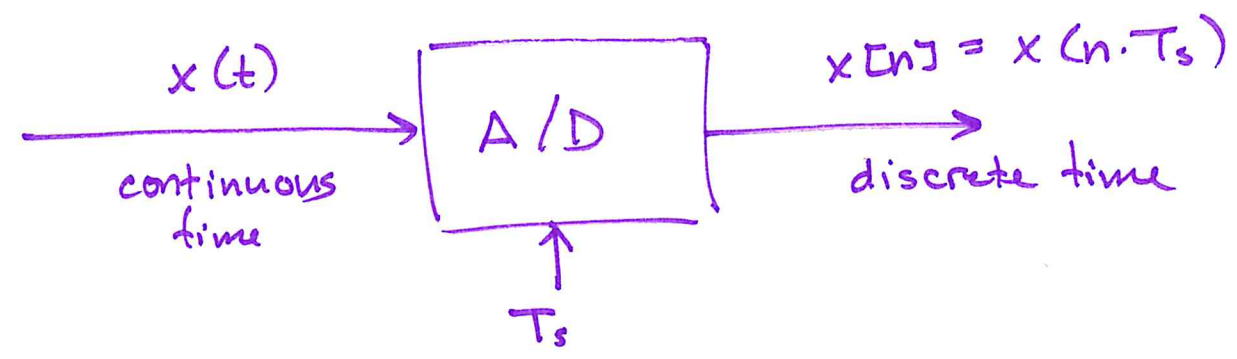
for $n = \dots, -2, -1, 0, 1, 2, \dots$

Ex CD Audio:

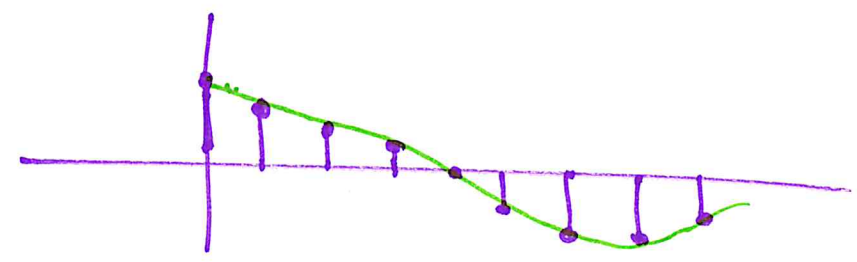
$$f_s = 44.1 \text{ kHz} \Rightarrow \begin{aligned} &44,100 \text{ samples/sec} \\ &\times 16 \text{ bits/sample} \leftarrow \\ &\times 2 \text{ channels} \\ &\times 60 \text{ sec/min} \\ &\times 3 \text{ min} \end{aligned}$$

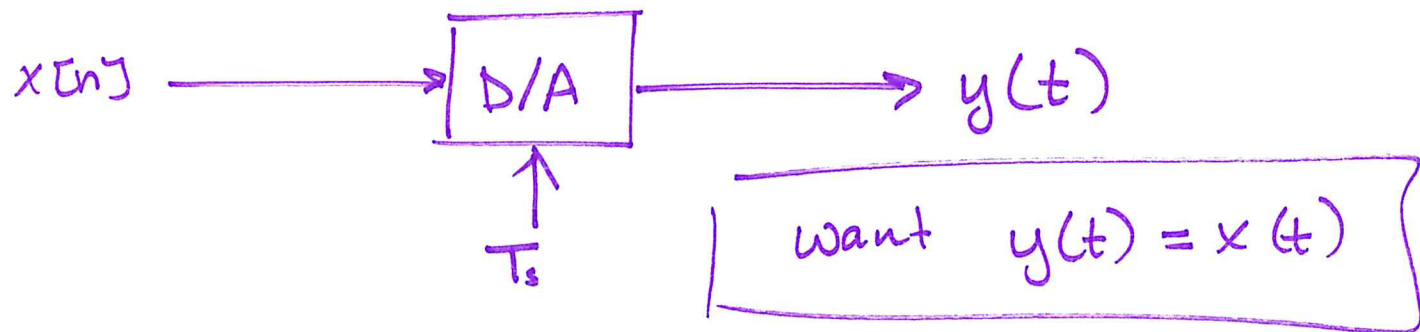
$$\approx 254 \text{ Mbits} = 32 \text{ MB}$$

Block diagram

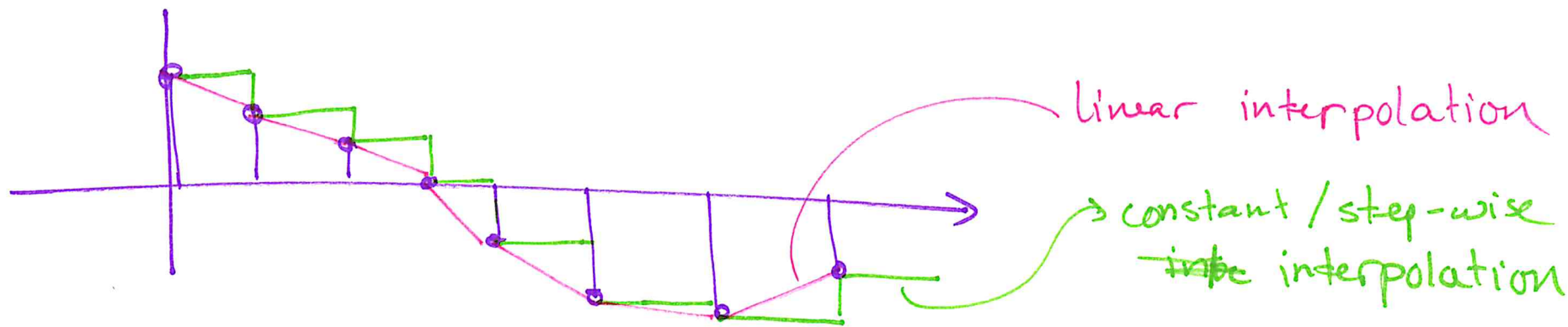


Also need to go from discrete time to continuous time
 "connecting the dots"





How do we connect the dots? / "Signal reconstruction"



errors between $x(t)$ and $y(t)$ get smaller as T_s gets smaller — BUT small $T_s \rightarrow$ more bits \rightarrow more storage \rightarrow requires more \$