

Fourier Transform

Part II

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Repetition:

$$\mathcal{F}[f(t)] = F(s) = \int_{-\infty}^{\infty} f(t) e^{-j2\pi st} dt$$

inverse:

$$\mathcal{F}^{-1}(F(s)) = f(t) = \int_{-\infty}^{\infty} F(s) e^{j2\pi st} ds$$

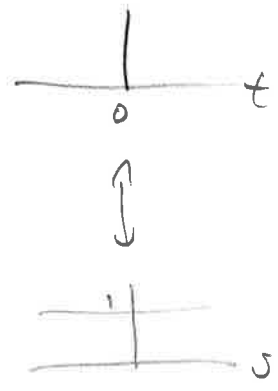
$$f(t) = \mathcal{F}^{-1}(\mathcal{F}(f(t))) \quad \text{Fourier Integral Theorem}$$

Example: a) impulse at origin;

$$F(s) = \int_{-\infty}^{\infty} \delta(t) e^{-j2\pi st} dt$$

sifting

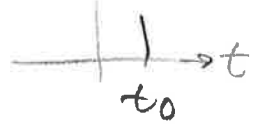
$$e^{-j2\pi s \phi} = e^{\phi} = 1$$



b) impulse at $t=t_0$

$$F(s) = \int e^{-j2\pi st} \delta(t-t_0) dt$$

$$= e^{-j2\pi s t_0} = \cos(2\pi s t_0) - j \sin(2\pi s t_0)$$



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②

last time:

$$a) \mathcal{F}(\cos(2\pi s_0 t)) = \dots = \frac{1}{2} [\delta(t-t_0) + \delta(t+t_0)]$$



solution not so trivial? in continuous domain

see before:

$$\begin{aligned} \delta(t-t_0) &\xrightarrow{\mathcal{F}} \cos(2\pi s t_0) + j \sin(2\pi s t_0) \\ \delta(t+t_0) &\xrightarrow{\mathcal{F}} \cos(2\pi s t_0) - j \sin(2\pi s t_0) \end{aligned}$$

\mathcal{F}^{-1}

use integral theorem

$$b) \mathcal{F}(\sin(2\pi s_0 t)) = \dots = \frac{1}{2j} (-\delta(t-t_0) + \delta(t+t_0))$$



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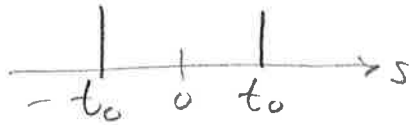
3

a word about Re/Im
and amplitude/phase :

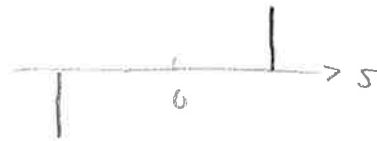
cosine

sine

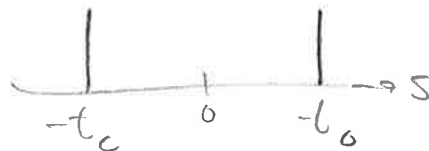
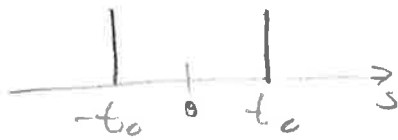
real



imaginary



amplitude



phase

\emptyset

$\pi/2$

Negative frequencies?

$$\cos(-f) = \cos(f)$$

$$\sin(-f) = -\sin(f)$$

} negative part is redundant*

=> after: we only plot the positive frequency components

* provide no additional information

Example

Gaussian

$$\left(\pi = \frac{1}{2\sigma^2} \Rightarrow \sigma^2 = \frac{1}{2\pi}, \sigma = \frac{1}{\sqrt{2\pi}} \right)$$

10/08/14
(4)

$$f(t) = e^{-\pi t^2}$$

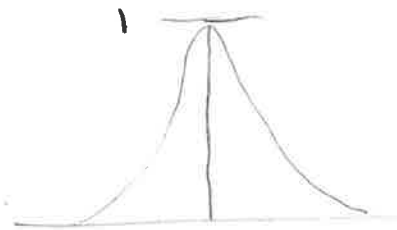
$$F(s) = \int_{-\infty}^{\infty} e^{-\pi t^2} e^{-j 2\pi s t} dt = \int_{-\infty}^{\infty} e^{-\pi (t^2 + j 2st)} dt$$

$$= \int_{-\infty}^{\infty} e^{-\pi (t + js)^2} \cdot e^{-\pi s^2} dt$$

$$\underbrace{t + js}_u$$

$$\begin{aligned} (t + js)^2 &= t^2 + 2tjs + j^2 s^2 \\ &= t^2 + 2j'st - s^2 \end{aligned}$$

$$= e^{-\pi s^2} \int_{-\infty}^{\infty} e^{-\pi u^2} du = e^{-\pi s^2}$$



general: $f(x) = \sqrt{\frac{\pi}{a}} e^{-\pi^2 x^2/a}$ || Large $\sigma \rightarrow$ small a
 \rightarrow wide in $F(s)$

normalized Gaussian: $g(x) = \frac{1}{\sqrt{2\pi}} e^{-\frac{1}{2}(\frac{x-\mu}{\sigma})^2}$ | $\mu=0$
 $\sigma=1$

$$= \frac{1}{\sqrt{2\pi}} e^{-\frac{1}{2}x^2} = e^{-\pi x^2} = e^{-\frac{1}{2}x^2}$$

general Gaussian:

$$\mathcal{F}[e^{-\pi(at)^2}] = \frac{1}{a} e^{-\pi\left(\frac{s}{a}\right)^2}$$

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(4)

↑
similarity theorem!
change of scale of abscissa

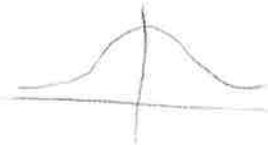
$$\left[\begin{aligned} F(s) &= \mathcal{F}[f(at)] = \int_{-\infty}^{\infty} f(at) e^{-i2\pi st} dt \\ &= \frac{1}{|a|} F\left(\frac{s}{a}\right) \end{aligned} \right]$$

$$\begin{aligned} f(t) &= e^{-\pi(at)^2} \\ &\downarrow \\ F(s) &= e^{-\pi\left(\frac{s}{a}\right)^2} \end{aligned}$$

large $a \rightarrow \leftarrow$
narrower Gaussian

\Rightarrow

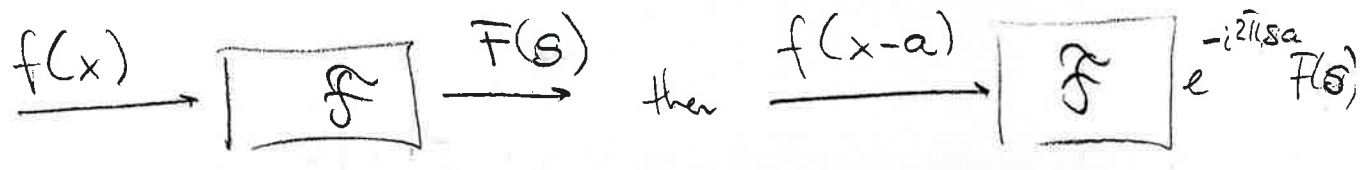
Frequency:



FT of convolution

(follow "Proof of ...")

Part I: shift invariance:



shifting $(x-a)$ does not change spectrum $F(u)$ but adds linear phase (mult by $e^{-j2\pi ua}$)
 Multiplying $F(u)$ by $e^{j2\pi ua}$ for different a translates / shifts $f(x)$ by a .

$$x' = x - a \quad dx = dx'$$

$$F[f(x-a)] = F[f(x')] = \int_{-\infty}^{\infty} f(x') e^{-j2\pi s(x'+a)} dx'$$

$$\left(e^{-j2\pi s(x'+a)} = \underbrace{e^{-j2\pi sa}}_{\text{constant}} e^{-j2\pi s x'} \right)$$

$$\Rightarrow F[f(x-a)] = e^{-j2\pi sa} F(s)$$

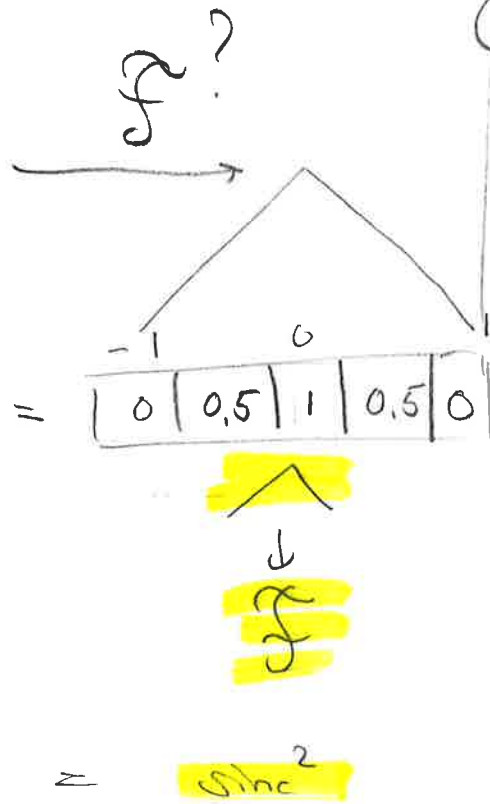
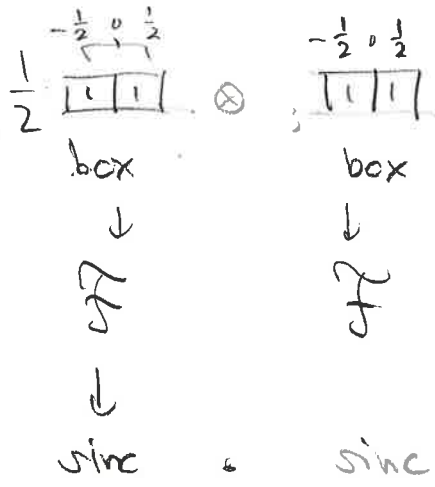
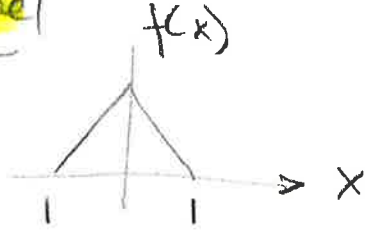
\Rightarrow FT is shift invariant

show deriv

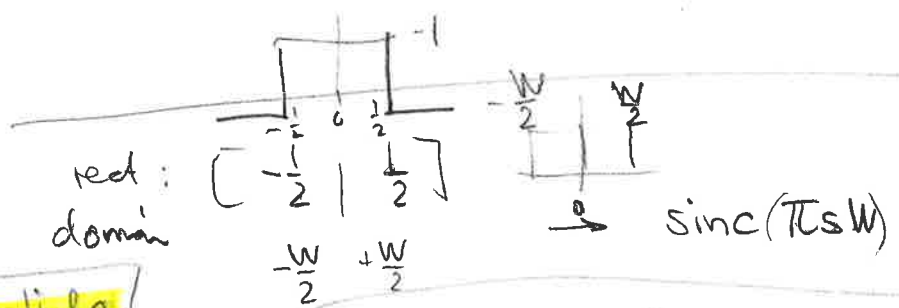
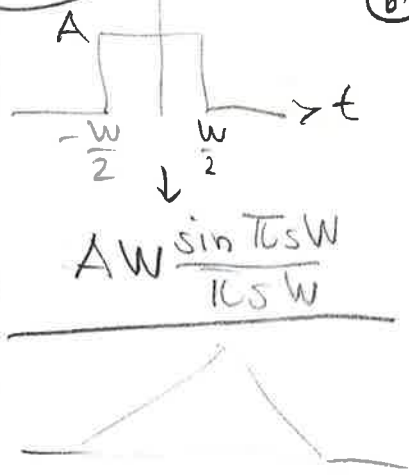
* Adds linear phase $\theta = sa$ to original phase.
 Or: adding linear phase filter produces transl. of signal.

Quiz

see slide



notes 10/08/14 (6)

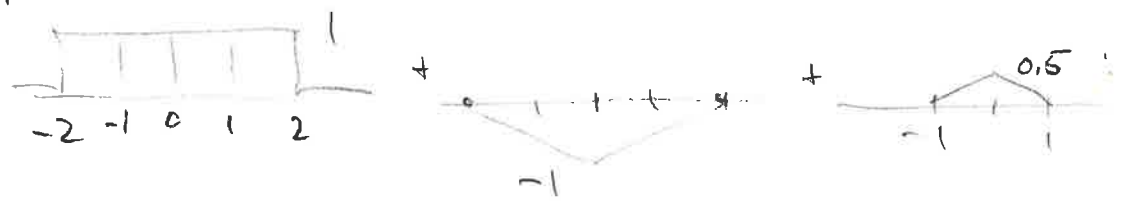
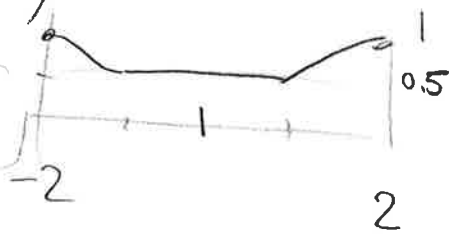
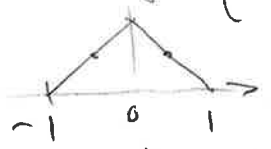


rect(x) $\begin{cases} 1 & \frac{x}{2} < x < \frac{x}{2} \\ 0 & \text{otherwise} \end{cases}$

see slide

\Rightarrow scaled to $[-2, 2]$ $\Rightarrow \frac{x}{4}$ ($\frac{2}{4} = \frac{1}{2}$)

triangle: $[-1, 1]$ scaled to $[-2, 2]$ $\Rightarrow \frac{x}{2}$



$f(x) = \text{rect}(\frac{x}{4}) - \Delta(\frac{x}{2}) + 0.5 \Delta(\frac{x}{2})$

$\Rightarrow F(s) = 4 \text{sinc}(4s) - 2 \text{sinc}^2(2s) + 0.5 \text{sinc}^2(s)$

FT Properties

Linear (shift invariant) filters: LSI (see ess 522)

$$x(t) \rightarrow \boxed{T} \rightarrow y(t) \quad | \quad y(t) = T[x(t)]$$

- ① shift invariant: $T[x(t+a)] = y(t+a)$
- ② scale invariant: $T[ax(t)] = ay(t)$
- ③ superposition invariant: $T[x_1(t) + x_2(t)] = T[x_1(t)] + T[x_2(t)]$

Convolution

$$c(t) = a(t) \otimes b(t) = \int_{-\infty}^{\infty} a(\tau) b(t-\tau) d\tau$$

- commutative: $a(t) \otimes b(t) = b(t) \otimes a(t)$

- associative: $(a(t) \otimes b(t)) \otimes c(t) = a(t) \otimes (b(t) \otimes c(t))$

- distributive: $a(t) \otimes [b(t) + c(t)] = [a(t) \otimes b(t)] + [a(t) \otimes c(t)]$

sum of 2 convolved signals
= convolution of sum

- Shift invariant: $\begin{array}{ccc} & T & \\ & \downarrow & \\ x(t+a) & \longrightarrow & y(t+a) \end{array}$
 produces shifted output when give shifted input

Part II: Convolution Theorem

(14)
follows "Proof of..."

10/08/14

$$f(t) \otimes g(t) = \int_{-\infty}^{\infty} f(\tau) g(t-\tau) d\tau$$

(7)

\mathcal{F} -transform:

$$\mathcal{F}[f \otimes g] = \int_{-\infty}^{\infty} \left[\int_{-\infty}^{\infty} f(\tau) g(t-\tau) d\tau \right] e^{-j2\pi st} dt$$

function of t
after evaluation of \int over τ
reverse order of integration

$$\int_{-\infty}^{\infty} f(\tau) \left[\int_{-\infty}^{\infty} g(t-\tau) e^{-j2\pi st} dt \right] d\tau$$

($f(\tau)$ can be treated as constant when integrating over dt)
- move bracket

Use Part I: $\int g(t-\tau) e^{-j2\pi st} dt = \mathcal{F}[g(t-\tau)] = e^{-j2\pi s\tau} G(s)$

$$\int_{-\infty}^{\infty} f(\tau) e^{-j2\pi s\tau} G(s) d\tau$$

$$= \left[\int_{-\infty}^{\infty} f(\tau) e^{-j2\pi s\tau} d\tau \right] G(s)$$

$$= F(s) G(s) \Rightarrow \mathcal{F}[f \otimes g] = F(s) G(s)$$

Convolution in space/time domain is equivalent to multiplication in the frequency domain.

$F(s) \otimes G(s)$
convolut-
in frequen-
space

$f(t) \otimes g(t)$
mult in
time/space domain

Symmetry

Inverse filtering, Reconstruction

$f(t)$: filter

$g(t)$: signal

$h(t)$: filtered signal

$$H(s) = \overbrace{F(s)G(s)}^{\text{measured}}$$

↑
known

$$G(s) = \frac{H(s)}{F(s)} \quad | F(s) \neq 0$$

$$g(t) = \mathcal{F}^{-1} \left[\frac{H(s)}{F(s)} \right]$$

example: image is corrupted by blurring (out of focus)

=> sharpening of image?

✓ filter is known (e.g. Gaussian)

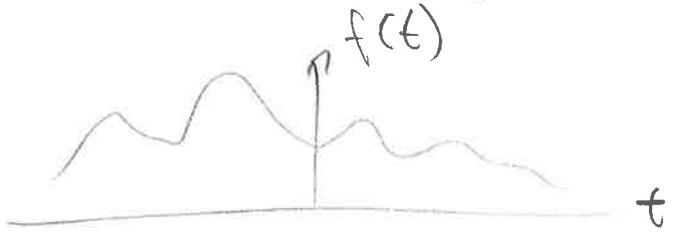
✓ image $f(t) \otimes g(t)$ is acquired

↓
estimate $g(t)$

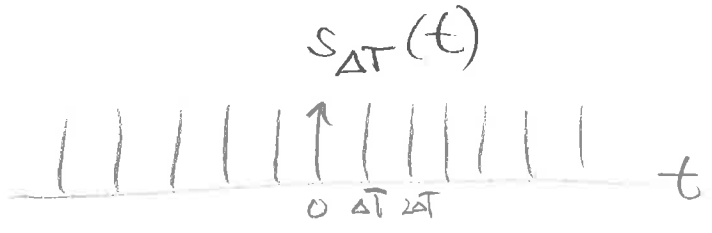
not so simple, but idea is right. ▽

4.3.1 Sampling

(DIP 4.31, page 211/212)



(e.g. camera sensor, scanner;
signal measured at discrete locations)



train of impulses

$$s_{\Delta T}(t) = \sum_{k=-\infty}^{\infty} \delta(t - k\Delta T)$$



sampled values

→ Sifting property (4.2.3)

sampling: $\tilde{f}(t) = f(t) s_{\Delta T}(t) = \sum_{n=-\infty}^{\infty} f(t) \delta(t - n\Delta T)$

multiply signal $f(t)$
by sampling function

$$\Rightarrow f_k = \int_{-\infty}^{\infty} f(t) \delta(t - k\Delta T) dt = f(k\Delta T)$$

sifting property:
(value of $f(t)$ at $t = k\Delta T$)

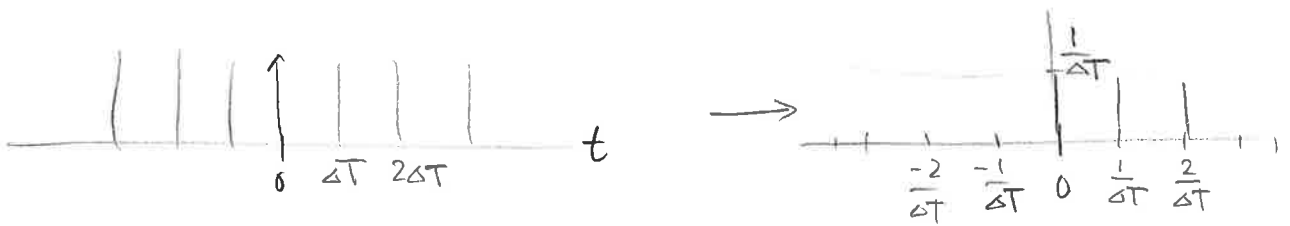
4.3.2 Fourier Transform of Sampled Function

$$F(s) = \mathcal{F}(\overset{\text{sampled}}{\tilde{f}(t)}) = \mathcal{F}(f(t) s_{\Delta T}(t))$$

convolution theorem

$$\Downarrow F(s) \otimes S(s)$$

- $F(s) : \mathcal{F}(f(t))$
- $S(s) : \frac{1}{\Delta T} \sum_{n=-\infty}^{\infty} \delta(s - \frac{n}{\Delta T})$ (see p.208/209)



(book 213) $\Rightarrow \boxed{F \sim} = F(s) \otimes S(s) = \int_{-\infty}^{\infty} F(\tau) S(s - \tau) d\tau$

$$= \frac{1}{\Delta T} \int_{-\infty}^{\infty} F(\tau) \sum_{n=-\infty}^{\infty} \delta(s - \tau - \frac{n}{\Delta T}) d\tau$$

$$= \frac{1}{\Delta T} \sum_{n=-\infty}^{\infty} \int_{-\infty}^{\infty} F(\tau) \delta(s - \tau - \frac{n}{\Delta T}) d\tau$$

$$= \frac{1}{\Delta T} \sum_{n=-\infty}^{\infty} F(s - \frac{n}{\Delta T})$$

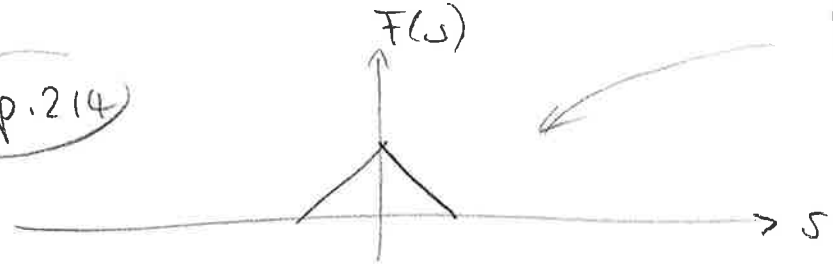
sifting property

What does that mean?

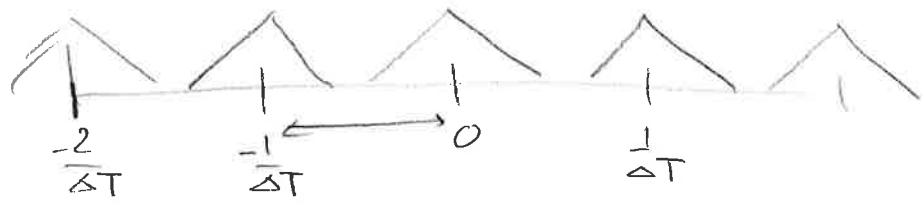
$\sim F(s)$: infinite, periodic sequence of copies of $F(s)$, with $F(s)$ transform of original, continuous function!

p.214

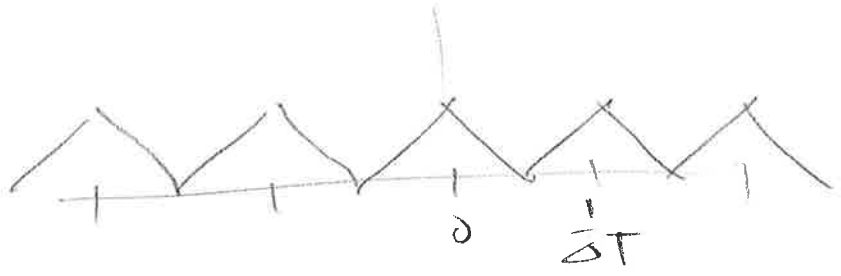
$F(f(t))$



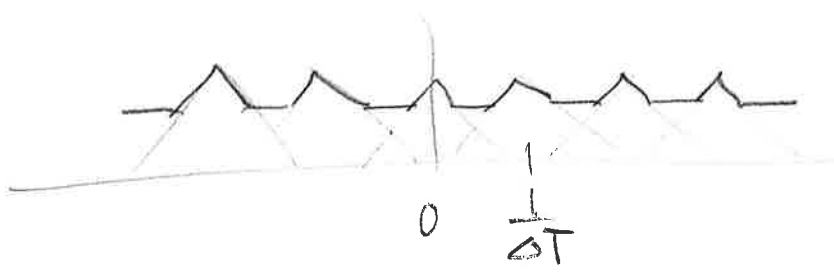
$\frac{1}{\Delta T}$: sampling rate
(cycles/mm, e.g.)



sample rate high enough to provide sufficient separation between periods; oversampled signal



just enough critically-sampled

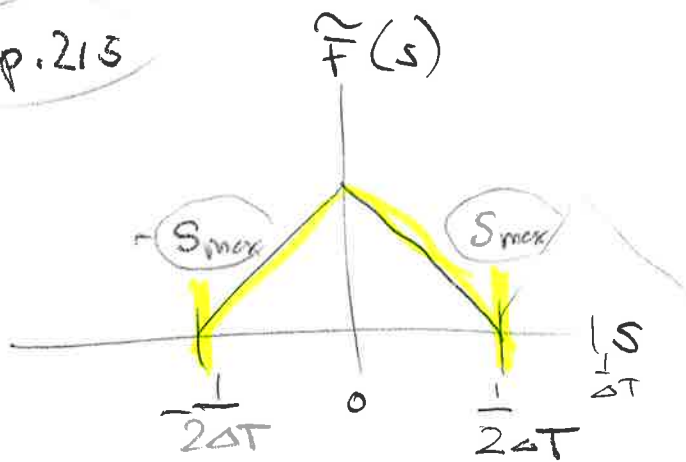


low value of $\frac{1}{\Delta T}$ causes periods to merge

coarse!
↓
sample rate below the minimum required to maintain distinguished copies of $F(s)$ → fails to preserve original transform; under-sampled signal

Recover ^{continuous} $f(t)$ from $\tilde{F}(s)$ (FT of sampled $f(t)$)

need one complete period to characterize entire transform!



$$\frac{1}{2\Delta T} > S_{\max}$$

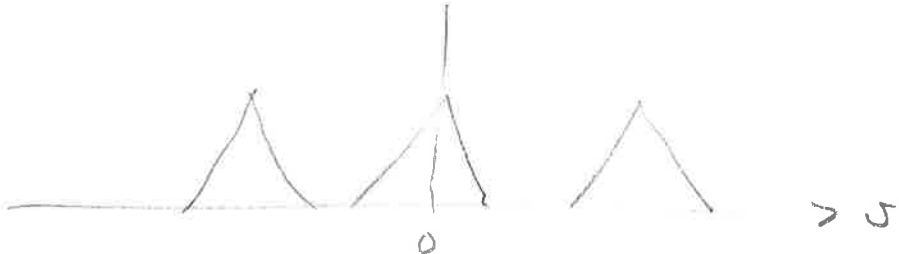
$$\frac{1}{\Delta T} > 2 S_{\max}$$

Sampling Theorem

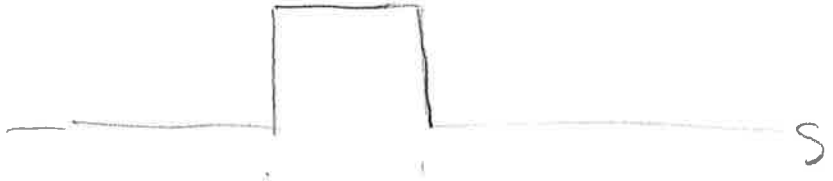
- Continuous, band-limited, function can be recovered completely from a set of its samples if samples acquired at a rate exceeding twice the highest frequency content of the function.
- Exactly twice the highest frequency: Nyquist rate
- Conversely:
Maximum frequency that can be "captured" by sampling a signal at a rate $\frac{1}{\Delta T}$ is $S_{\max} = \frac{1}{2\Delta T}$.

Procedure

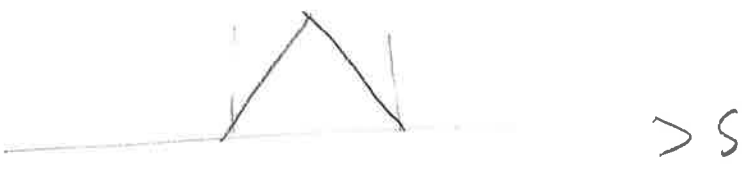
sampled
 $\tilde{F}(s)$



$\tilde{F}(s)$

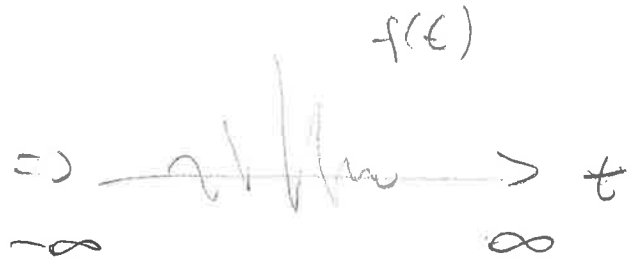


$H(s)$: low-pass filter



$F(s) = H(s)\tilde{F}(s)$

$\updownarrow \mathcal{F}^{-1}$



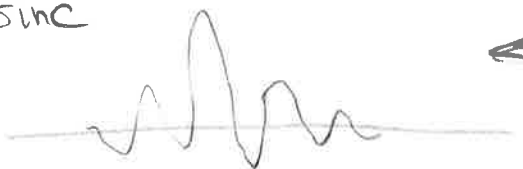
$f(t) = \mathcal{F}^{-1}(F(s))$

real world: $f(t)$ has limited range $\int_{-t_{max}}^{t_{max}}$

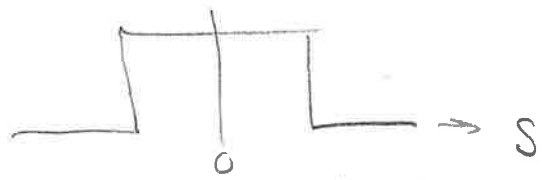
\Rightarrow ideal recovery impossible

rect function in frequency domain?

sinc



convolution with sinc
in space/time domain



cutting higher frequencies in frequency domain

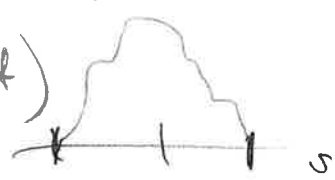
$$\Rightarrow \overset{\text{reconstructed}}{f(t)} = \overset{\text{sampled}}{\tilde{f}(t)} \otimes \text{sinc}\left(\frac{t}{\Delta T}\right)$$

$$= \sum_k \overset{\text{samples}}{f_k} \text{sinc}\left(\frac{t - k\Delta T}{\Delta T}\right)$$

See slide



Problem: most fcts not band limited!
 (power spectrum has finite support)



\Rightarrow forcing fcts to be band limited
 \rightarrow ringing artifacts

See slides

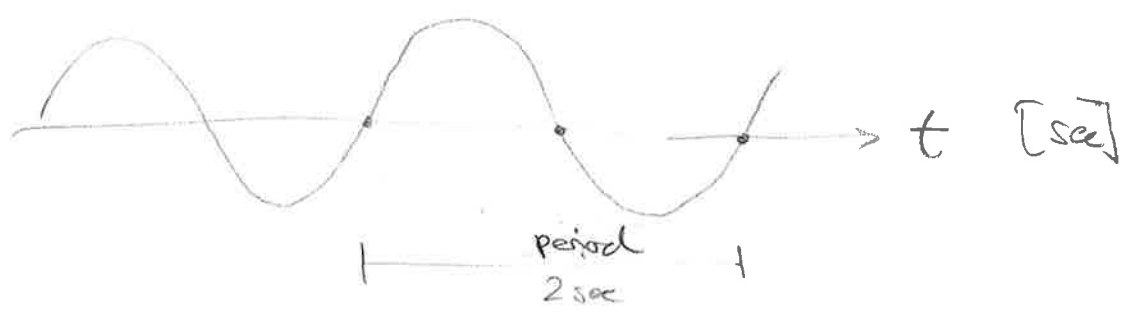
Aliasing 4.3.4

(Example book p. 219) sine wave \rightarrow single frequency \rightarrow band limited!

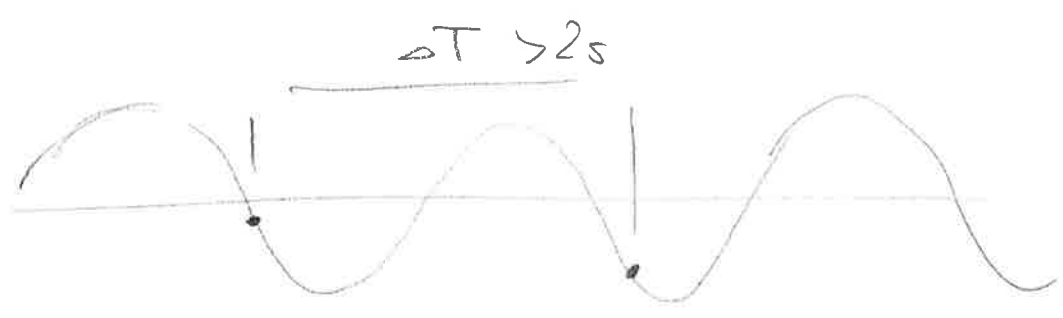
$\sin(\pi t)$ period is 2s, frequency $\frac{1}{2}$ cycles/sec

(sampling theorem: $\frac{1}{\Delta T} >$ twice highest frequency of signal
 $\frac{1}{\Delta T} > 1$ sample/sec

$\Delta T < 1s$



if exactly twice: $\Delta T = 1s \Rightarrow \sin(-\pi), \sin(0), \sin(\pi)$
 $\Rightarrow \Delta T < 1s$ all 0 ?



what if samples taken with separation $\Delta T > 2s$?

\Rightarrow sample rate $\frac{1}{\Delta T} < \frac{1}{2}$ samples/sec

\Rightarrow looks like sine wave, but frequency $\approx \frac{1}{10}$ Hz ?

\Rightarrow Aliasing