

Solution

1. (a) Linear System: one in which the response to a sum of inputs is sum of responses to individual inputs, i.e.

$$L \left[\sum_n \alpha_n x_n(t) \right] = \sum_n \alpha_n L[x_n(t)]$$

Implies additivity: $L[\sum x] = \sum L[x]$

+ homogeneous: $L(\alpha x) = \alpha L[x]$

- (b) Causal System: response comes after cause.
i.e. $\int_t^{\infty} x(\tau) h(t-\tau) d\tau = 0$

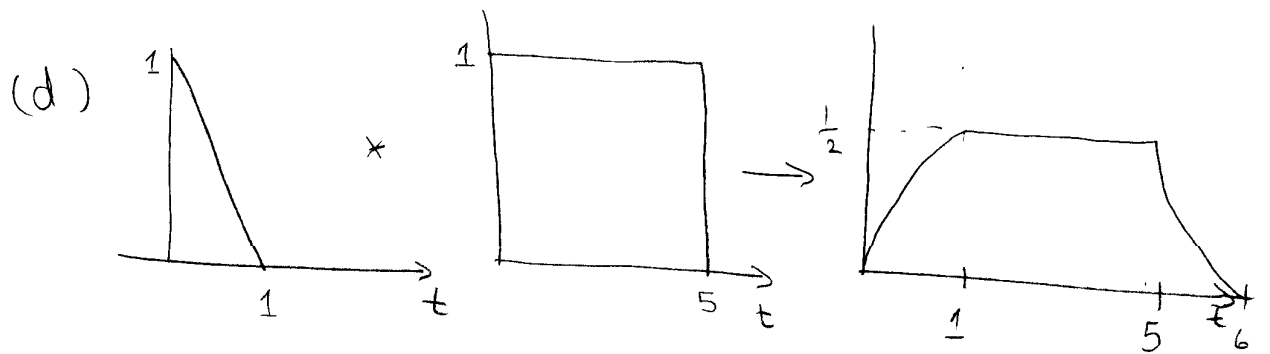
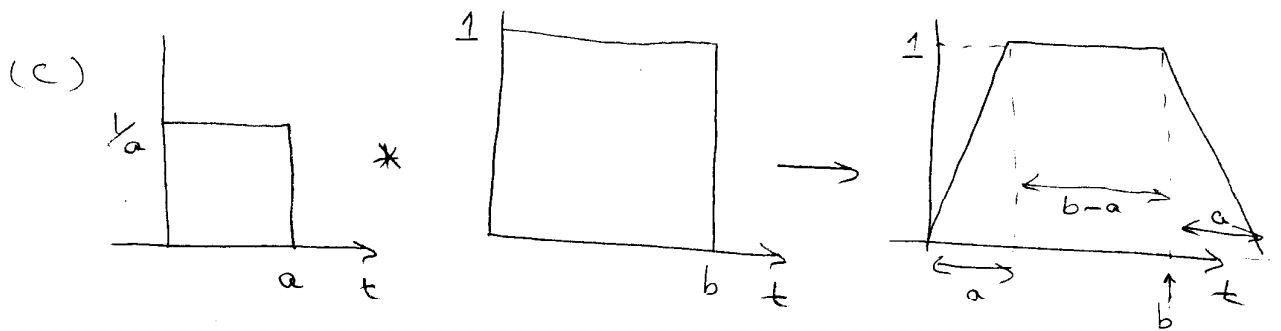
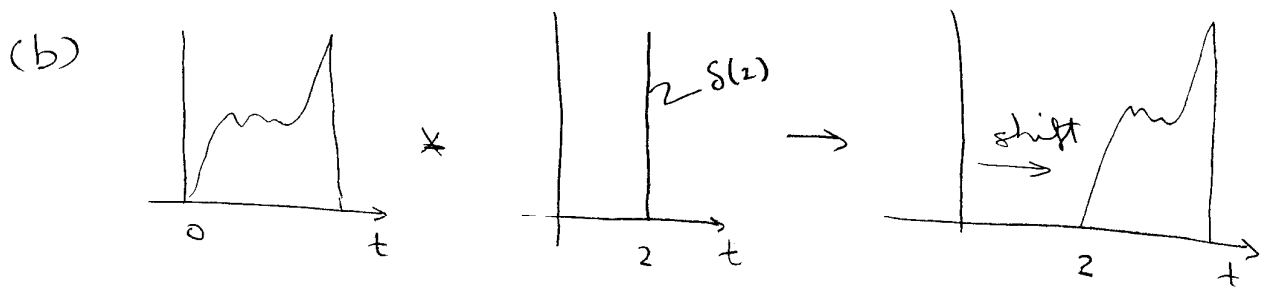
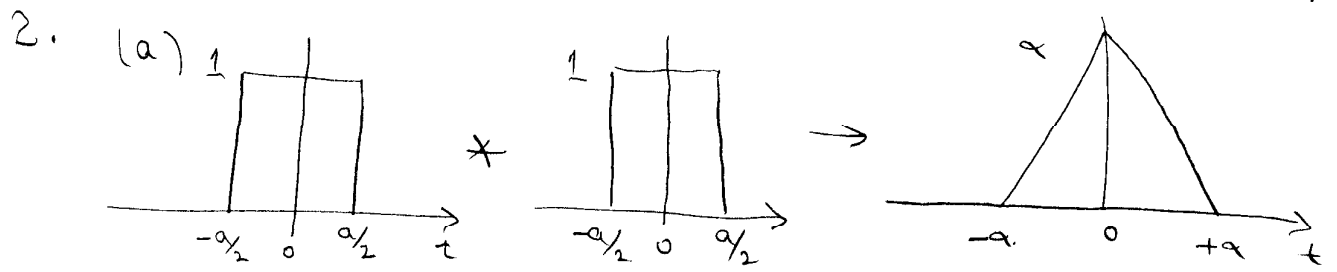
- (c) Time Invariant System: response to an input now is the same as the response to a later input.

- (d) Convolution Integral:

$\int_{-\infty}^{\infty} f(\tau) g(t-\tau) d\tau$, the response to a input in a linear system is the convolution of the input with the impulse response.

- (e) Autocorrelation:

$\int_{-\infty}^{\infty} f(\tau) f(t+\tau) d\tau$, a measure of the correlation of one part of a signal to another part of the same signal.



4. (a) When we reflect the time signal about the origin, the FT is even + real. This makes manipulation easier.

(b) We get a convolution integral when we make a measurement in a LTIC system because the response of an instrument (or any LTIC system in general) to an impulse is:

$$y(t) = \mathcal{L}[x(t)] = \int_{-\infty}^{\infty} x(\tau) \mathcal{L}[s(t-\tau)] d\tau \\ \equiv \int_{-\infty}^{\infty} x(\tau) h(t-\tau) d\tau$$

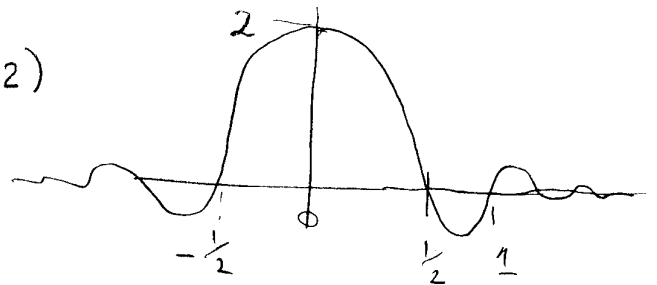
(c) FT's are useful for dealing with convolutions because, in ν space, it is just a multiplication, i.e. $f * g \Rightarrow F \cdot G$

(d) \mathcal{L} works in general due to the e^{-st} factor, i.e. \mathcal{L} has a real damping component. FT does not. So FT is okay for a real signal because it is finite in time + amplitude. We don't always have that in a differential equation so \mathcal{L} is used there. \mathcal{L} can also handle non-zero initial conditions. FT cannot.

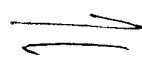
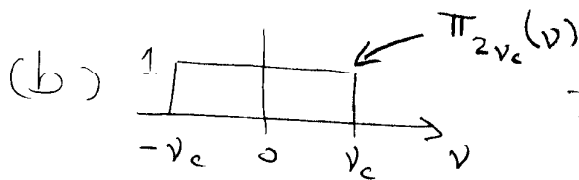
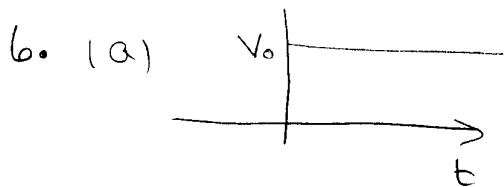
Also \mathcal{L}^{-1} is easily found by table lookup for known terms in a differential eqn. There is no known \mathcal{L}^{-1} for a general signal. There is for \mathcal{F}^{-1} .

5. (a) $[a * b + c d]^2 \iff [A \cdot B + C * D] * [A \cdot B + C * D]$
 where $a \iff A$, etc.

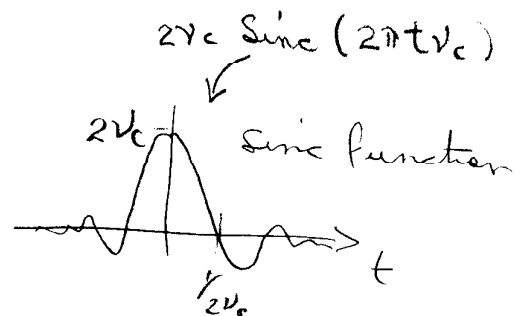
(b) $\Pi_2(t) \iff 2 \text{Sinc}(\pi v 2)$



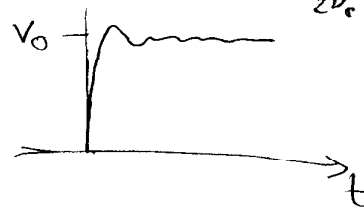
(c) $e^{-\pi t^2} \iff e^{-\pi v^2}$



(c)



(d) $y(t) = V_0 u(t) * \text{sinc}$



(e) $V_0 u(t) \iff V_0 \phi(v)$

where $\phi(v) = \frac{1}{2\pi i v}$ as per the lecture notes.

(f) $Y(v) = \underbrace{V_0}_{\text{input}} \underbrace{\frac{1}{2\pi i v} \Pi_{2v_c}(v)}_{\text{Filter}}$

$$g) \quad y(t) = \int_{-v_c}^{v_c} \frac{V_0}{2\pi i v} e^{2\pi i v t} dv$$

$$= V_0 t \int_{-v_c}^{v_c} \text{sinc } 2\pi vt \, dv \quad (\text{see lecture notes page 8-6})$$

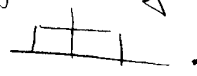
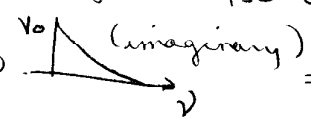

This needs to be numerically integrated & yields the same as (d).

h) Time space:

the input signal is convoluted with the sinc function (which is the ideal filter in t space)

Freq. space:

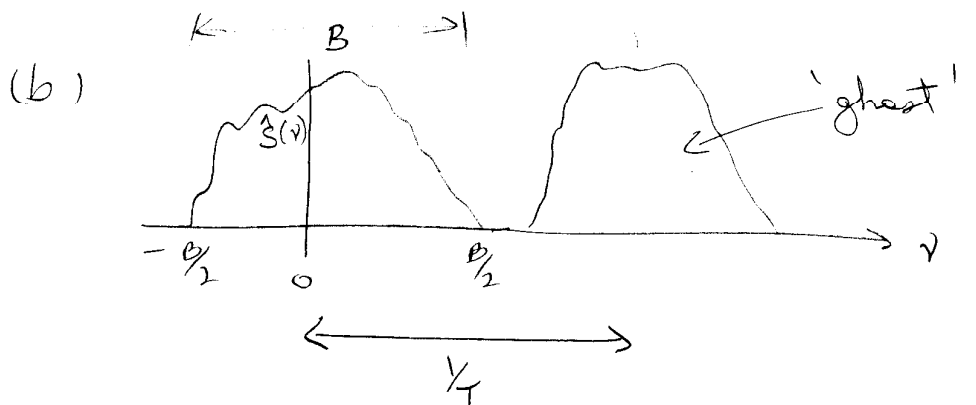
the high freq. components are filtered out

via . Thus  (imaginary) \Rightarrow 

7. (a) as per lecture notes:

$$\begin{aligned} \hat{S}\left(\nu + \frac{m}{T}\right) &= \sum_{n=-\infty}^{\infty} s(nT) e^{-2\pi i \nu nT - 2\pi i n m} \\ &= \sum_{n=-\infty}^{\infty} s(nT) e^{-2\pi i \nu nT} \cdot \underbrace{e^{-2\pi i n m}}_{=1} \\ &= \hat{S}(\nu) \end{aligned}$$

\therefore periodic QED since a multiple of 2π



The sampling rate $1/T$ must be large enough to prevent overlap of the ghost & the signal.

(c) Sampling rate ($1/T$) must be $\geq 2 \times$ highest frequency ($B/2$)

$$\text{ie } \frac{1}{T} \geq B$$

\therefore critical rate is $\frac{1}{T} = B$ ie, $T = \frac{1}{B}$

— end —