

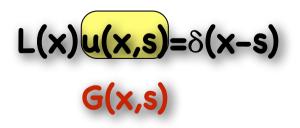


**Green's function** is a basic solution to a linear differential equation, a building block that can be used to construct many useful solutions.

If one considers a linear differential equation written as:

# L(x)u(x)=f(x)

where L(x) is a linear, self-adjoint differential operator, u(x) is the unknown function, and f(x) is a known nonhomogeneous term, the GF is a solution of:





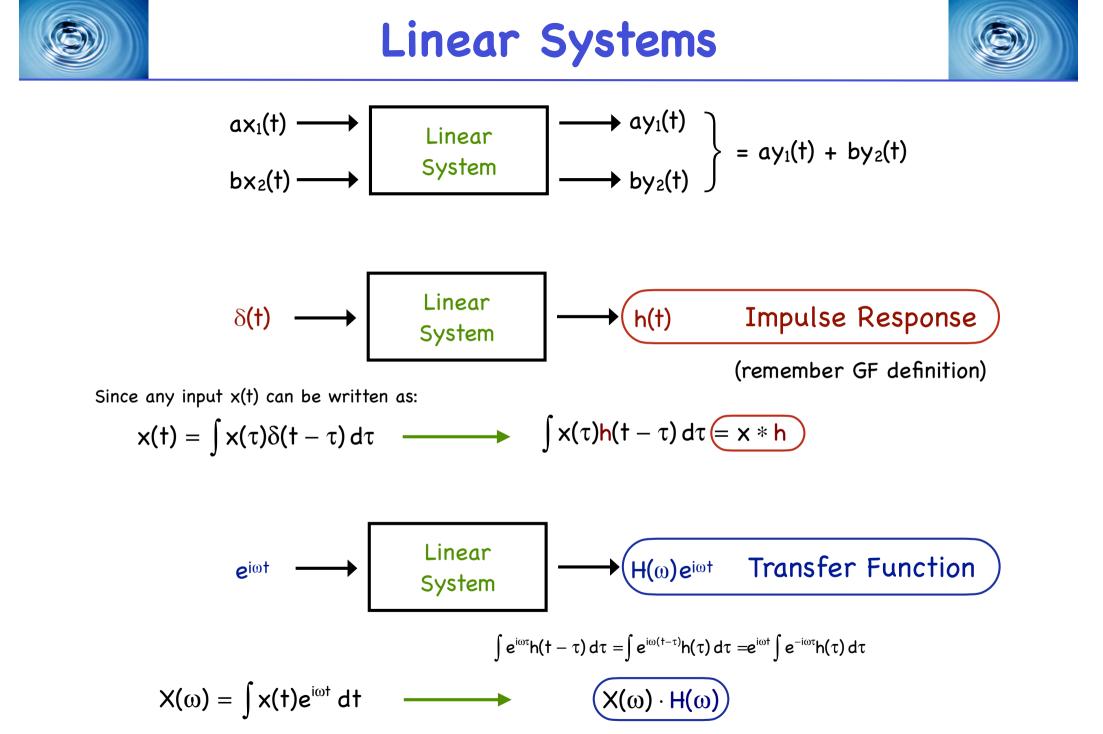


If such a function G can be found for the operator L, then if we multiply the second equation for the Green's function by f(s), and then perform an integration in the s variable, we obtain:

$$\int L(x)G(x,s)f(s)ds = \int \delta(x-s)f(s)ds = f(x) = Lu(x)$$
$$L\int G(x,s)f(s)ds = Lu(x)$$

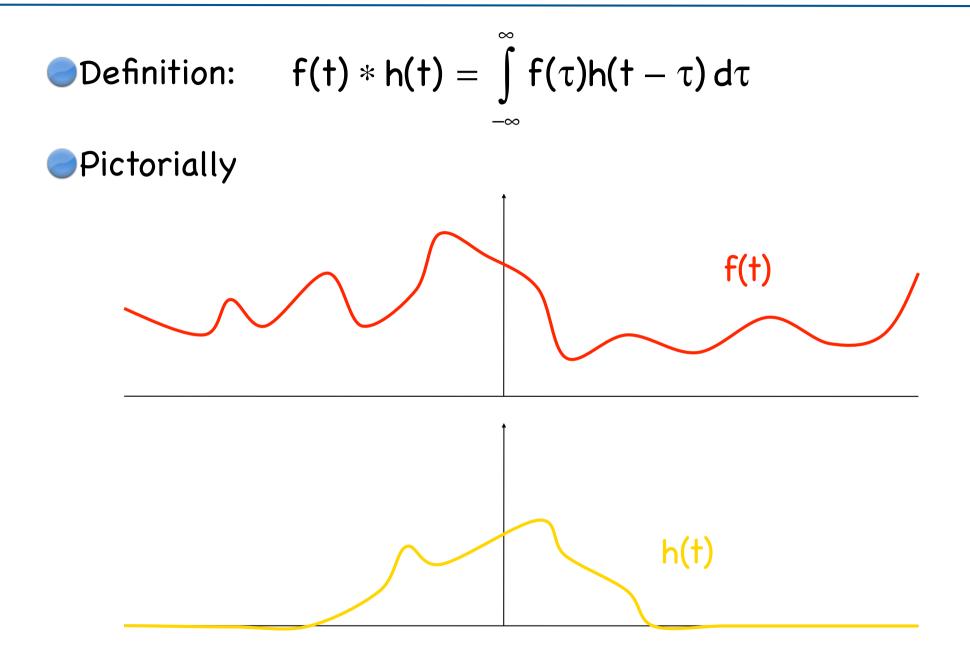
$$u(x) = \int G(x,s)f(s)ds$$

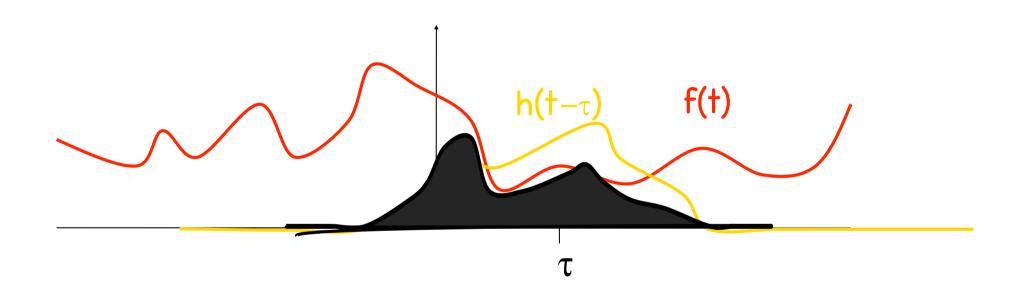
Thus, we can obtain the function u(x) through the knowledge of the Green's function and the source term. This process has resulted from the linearity of the operator L. See Linear System Theory (i.e. impulse response)





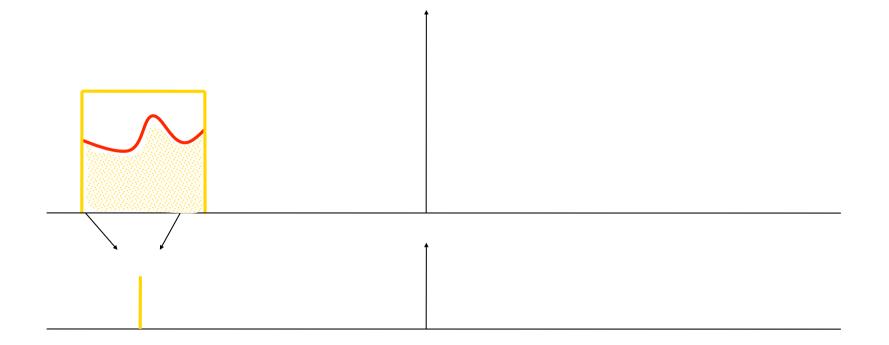
 $\infty$  $f(t) * h(t) = \int f(\tau)h(t-\tau) d\tau$  $-\infty$ 

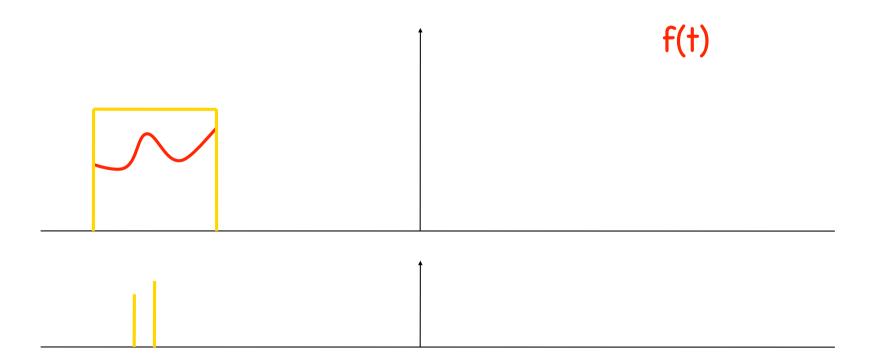


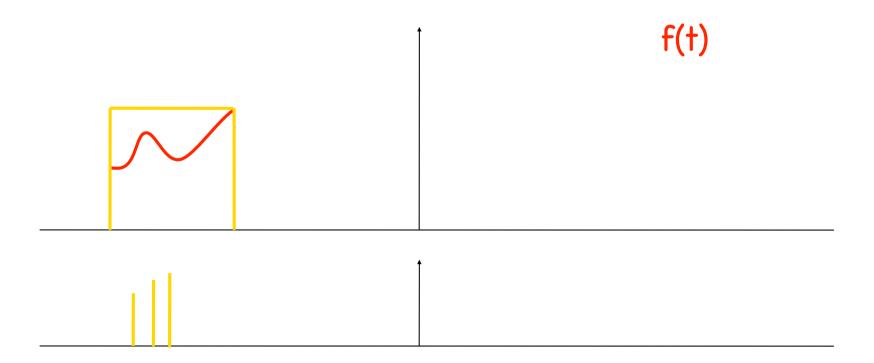


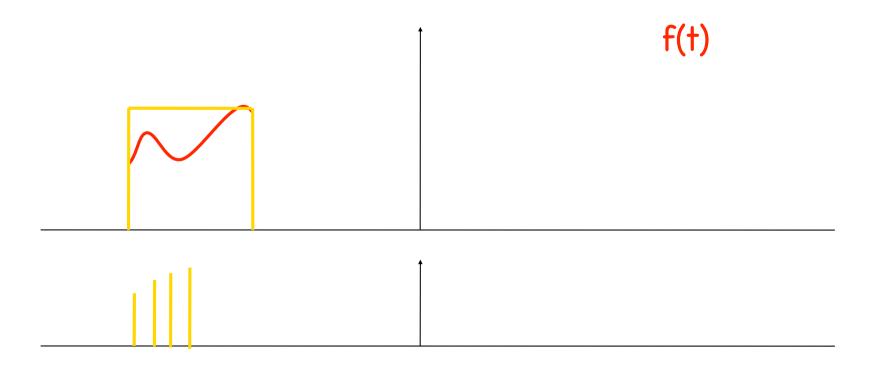
Consider the boxcar function (box filter):

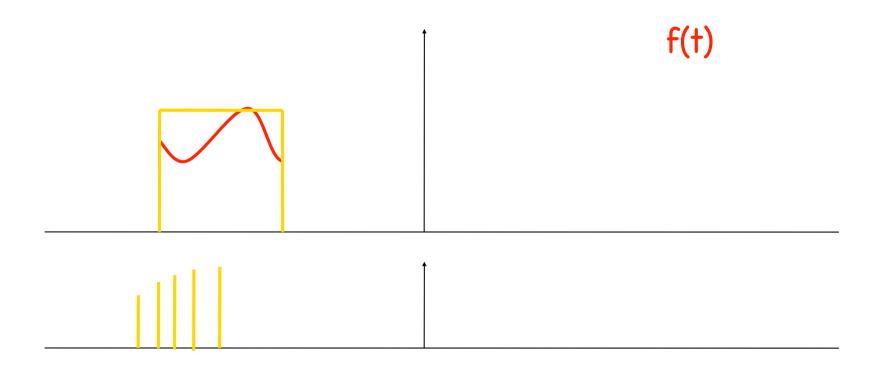
$$h(t) = \begin{cases} 0 & t < -\frac{1}{2} \\ 1 & -\frac{1}{2} \le t \le \frac{1}{2} \\ 0 & t > \frac{1}{2} \end{cases}$$

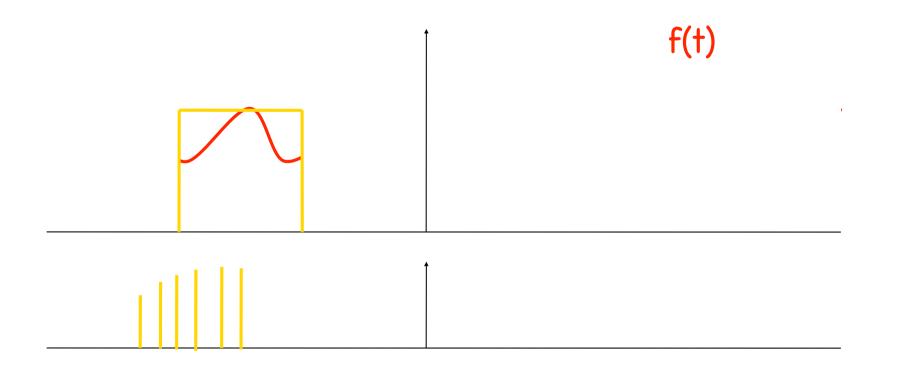


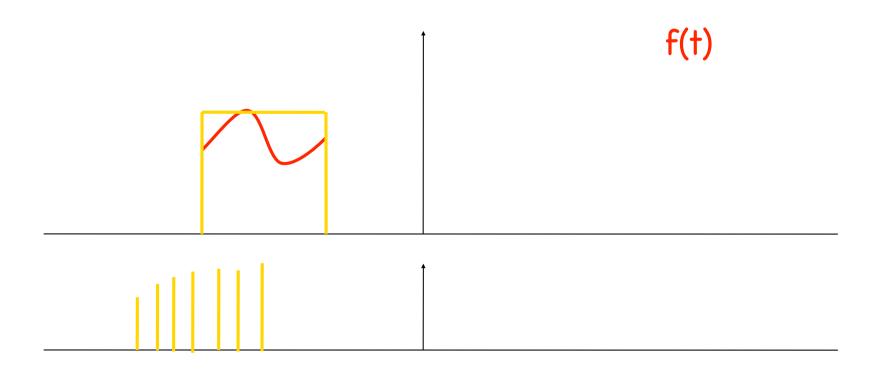


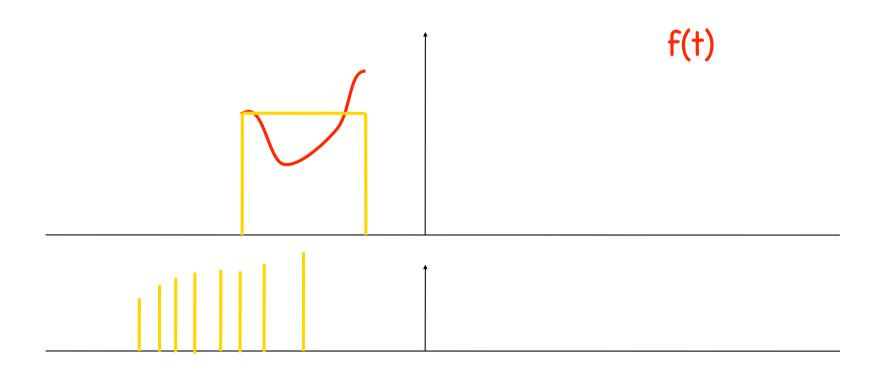


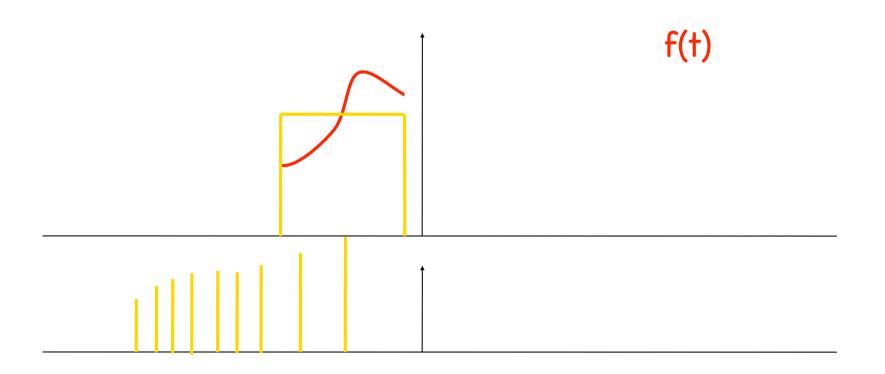


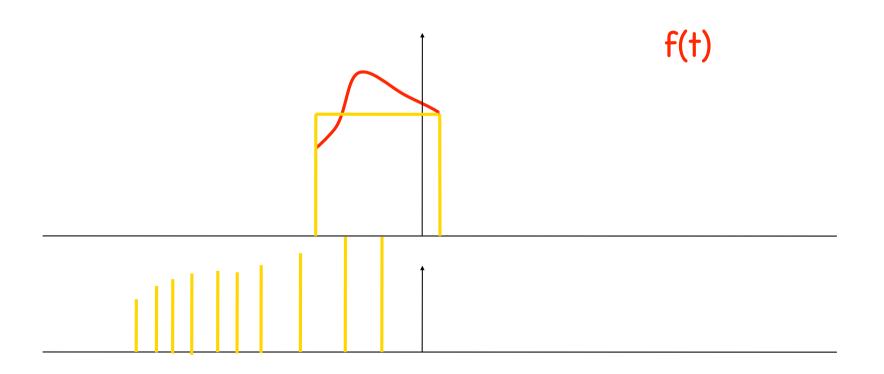


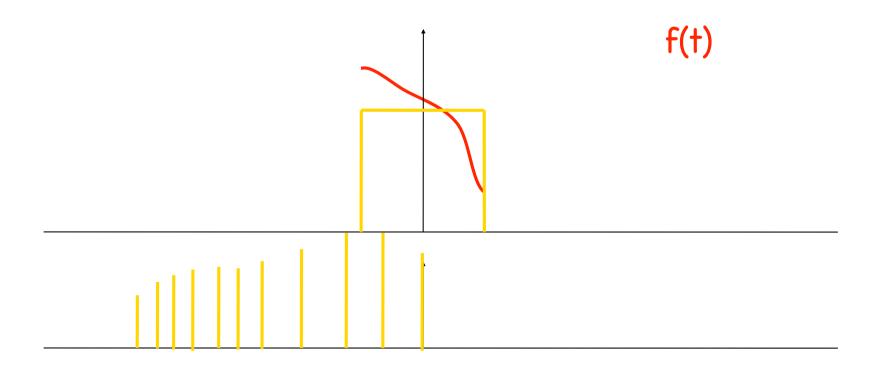


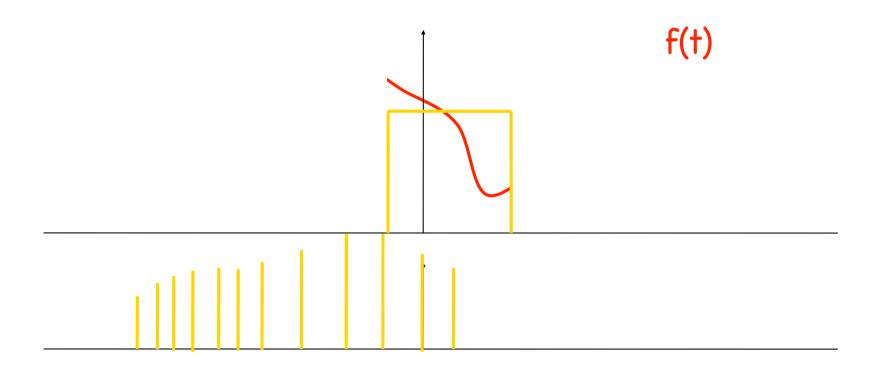


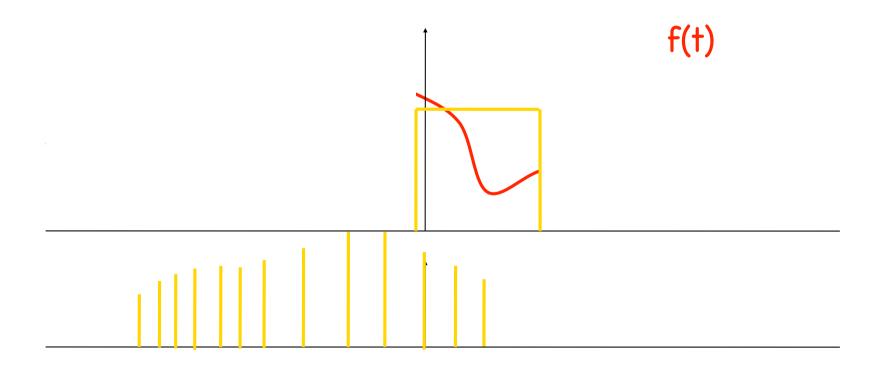


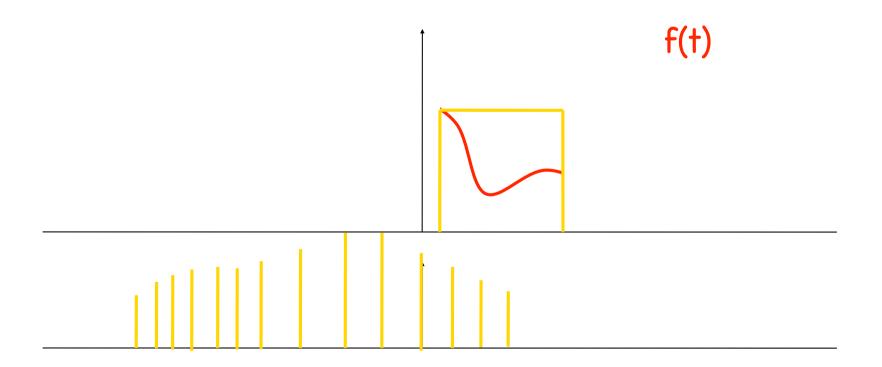


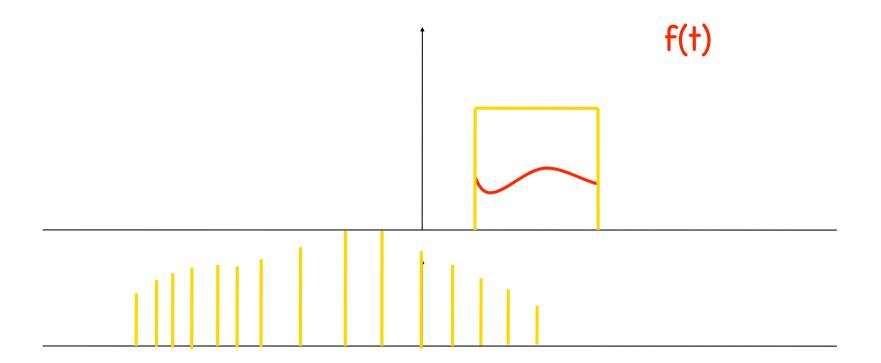


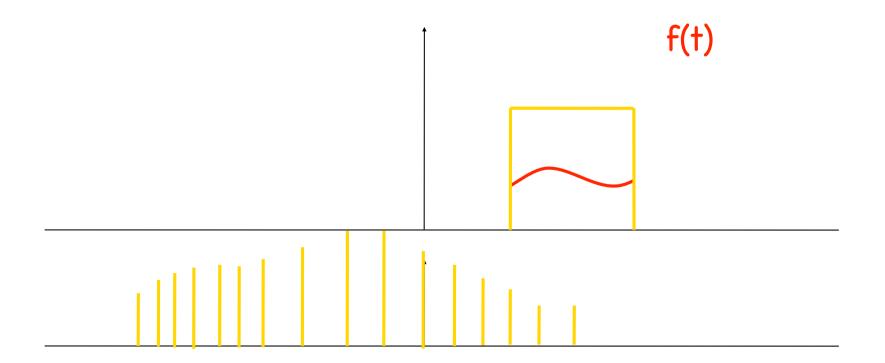


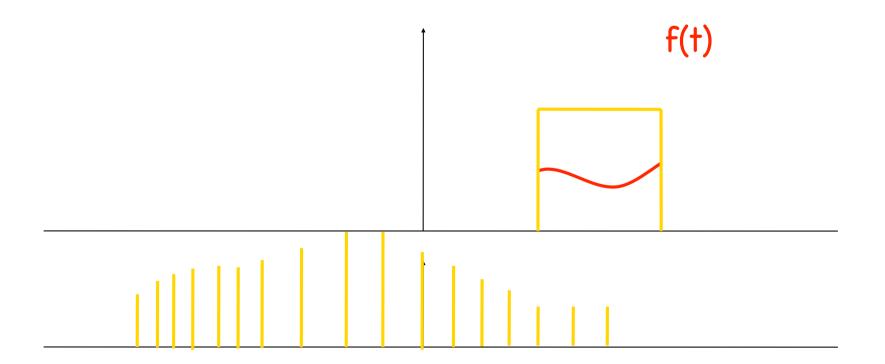


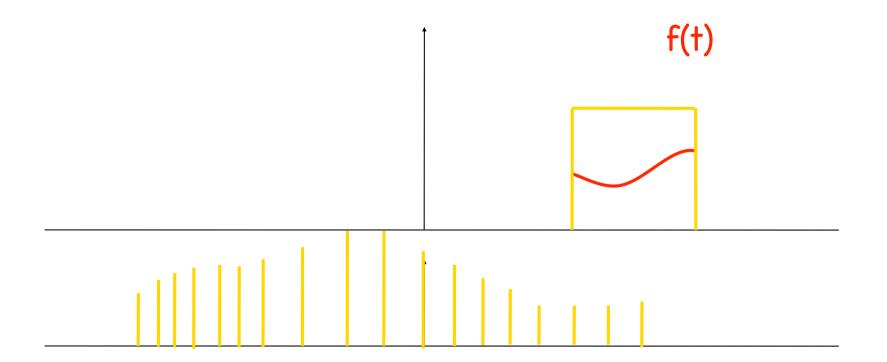


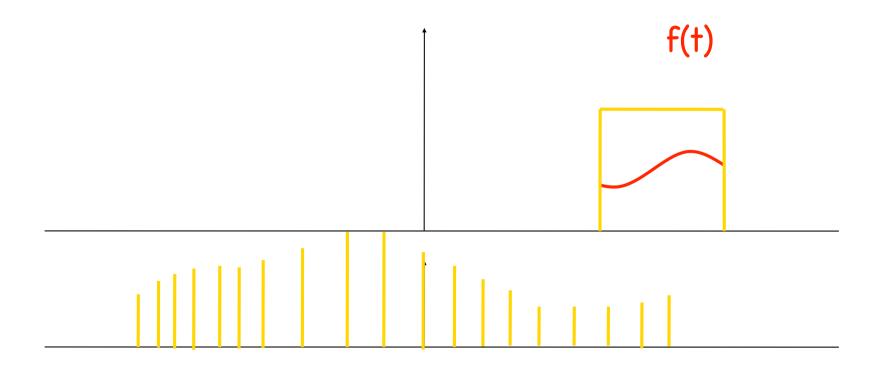


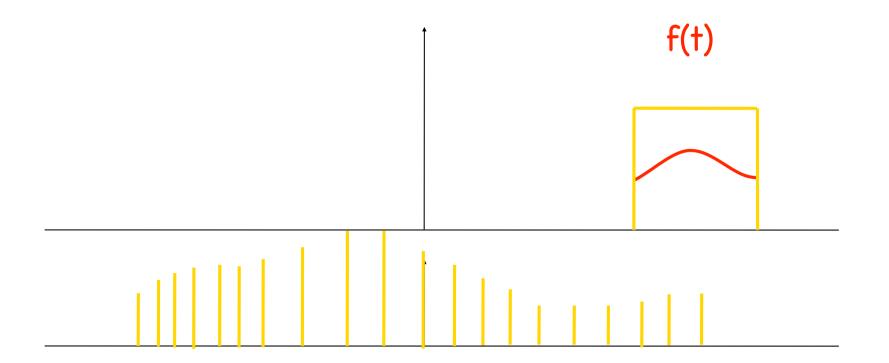


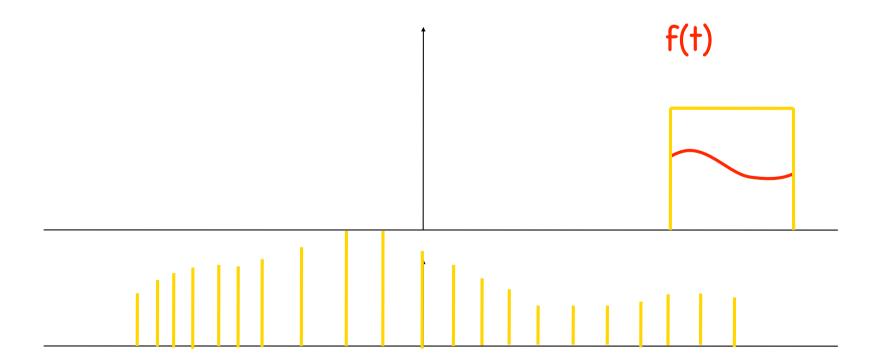


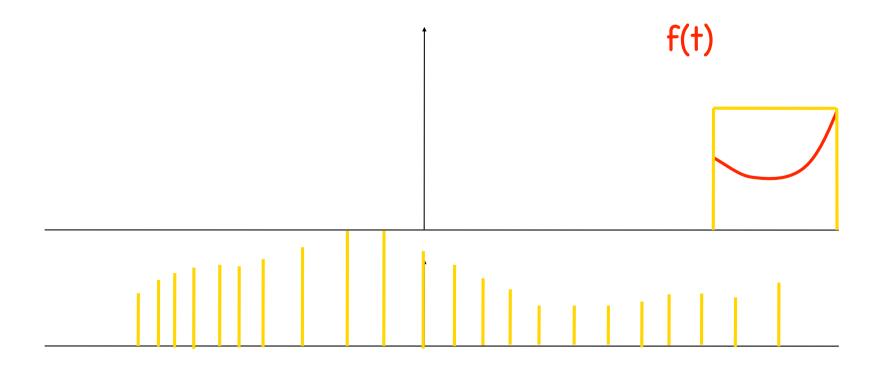




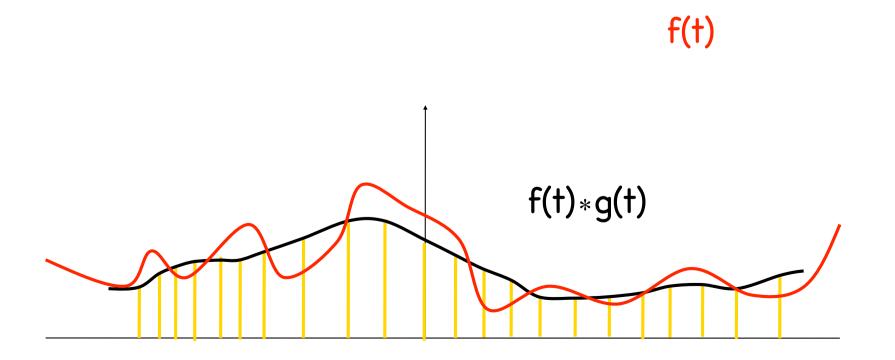








This particular convolution smooths out some of the high frequencies in f(t).

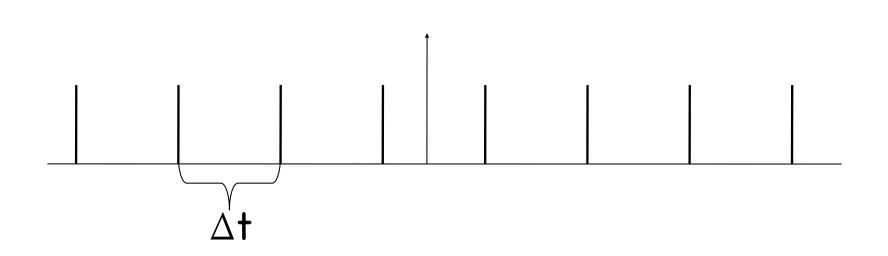


#### Sampling Function

A Sampling Function or Impulse Train is defined by:

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

where  $\Delta t$  is the sample spacing.



## Sampling Function

The Fourier Transform of the Sampling Function is itself a sampling function.

The sample spacing is the inverse.

 $S_{\Delta^{\dagger}}(\dagger) \Leftrightarrow S_{\frac{1}{M}}(\omega)$ 

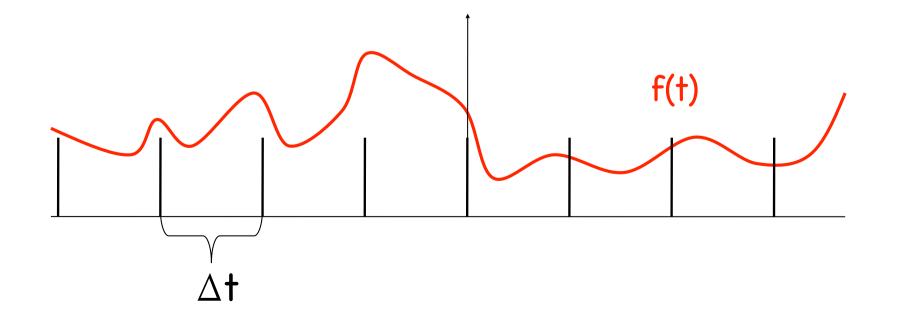
The convolution theorem states that convolution in the temporal domain is equivalent to multiplication in the frequency domain, and viceversa.

# $f(t) * g(t) \Leftrightarrow F(\omega) \cdot G(\omega)$

 $f(t) \cdot g(t) \Leftrightarrow F(\omega) * G(\omega)$ 

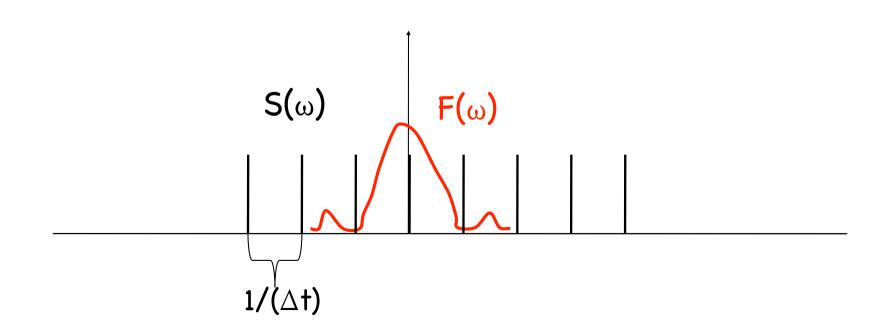
This powerful theorem can illustrate the problems with our point sampling and provide guidance on avoiding aliasing.

Consider:  $f(t) \cdot S_{\Delta t}(t)$ 



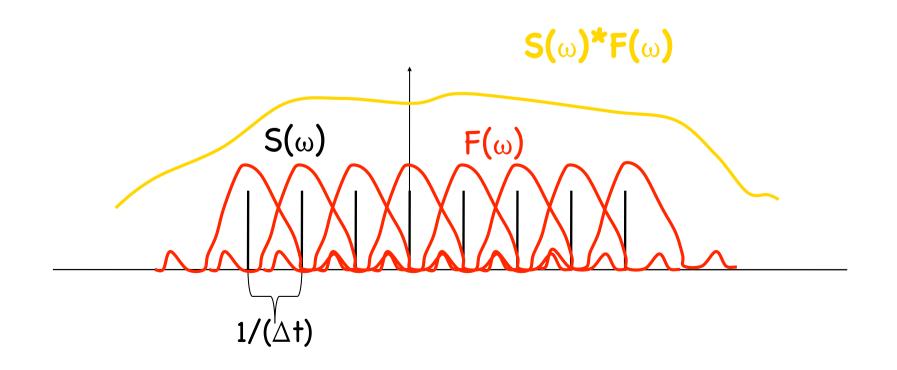
#### **Convolution Theorem**

What does this look like in the Fourier domain?



#### **Convolution Theorem**

In Fourier domain we would convolve



# Aliasing

What this says, is that any frequencies greater than a certain amount will appear intermixed with other frequencies.

In particular, the higher frequencies for the copy at  $1/\Delta t$  intermix with the low frequencies centered at the origin.

# Aliasing and Sampling

Note, that the sampling process introduces frequencies out to infinity.

- We have also lost the function f(t), and now have only the discrete samples.
- This brings us to our next powerful theory.

#### The Shannon Sampling Theorem:

A band-limited signal f(t), with a cutoff frequency of  $\lambda$ , that is sampled with a sampling spacing of  $\Delta t$ may be perfectly reconstructed from the discrete values f[n $\Delta t$ ] by convolution with the sinc(t) function, provided the Nyquist limit:  $\lambda < 1/(2\Delta t)$ 

Why is this?

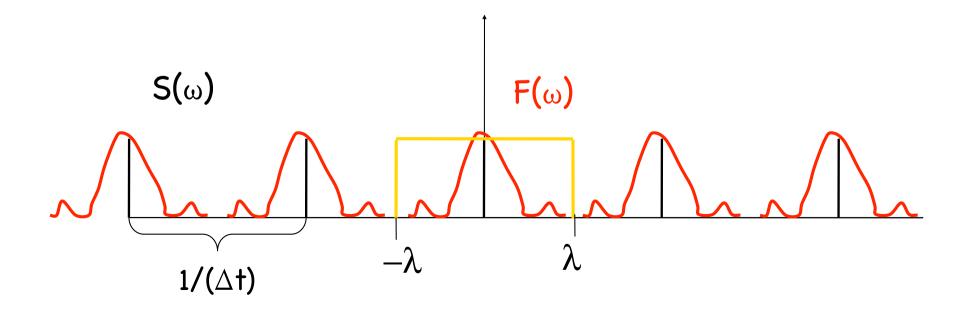
The Nyquist limit will ensure that the copies of  $F(\omega)$  do not overlap in the frequency domain.

We can completely reconstruct or determine f(t) from  $F(\omega)$  using the Inverse Fourier Transform.

# Sampling Theory

In order to do this, we need to remove all of the shifted copies of  $F(\omega)$  first.

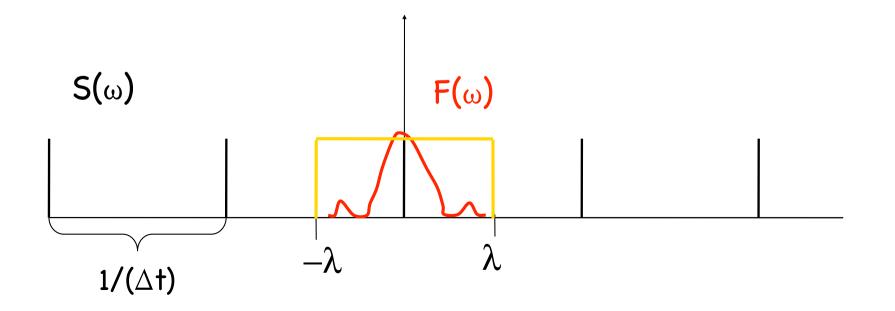
This is done by simply multiplying  $F(\omega)$  by a box function of width  $2\lambda$ .



# Sampling Theory

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This is done by simply multiplying  $F(\omega)$  by a box function of width  $2\lambda$ .



So, given  $f[n\Delta t]$  and an assumption that f(t) does not have frequencies greater than  $1/(2\Delta t)$ , we can write the formula:

 $f[nT] = f(t) \cdot S_{\Delta t}(t) \Leftrightarrow F(\omega) * S_{\Delta t}(\omega)$ 

 $\mathsf{F}(\omega) = (\mathsf{F}(\omega) * \mathsf{S}_{\Delta \dagger}(\omega)) \cdot \mathsf{Box}_{1/(2\Delta \dagger)}(\omega)$ 

therefore,

 $f(t) = f[n\Delta t] * sinc(t)$ 

http://www.thefouriertransform.com/pairs/box.php

http://195.134.76.37/applets/AppletNyquist/Appl\_Nyquist2.html