

Machine Learning in Imaging

BME 590L
Roarke Horstmeyer

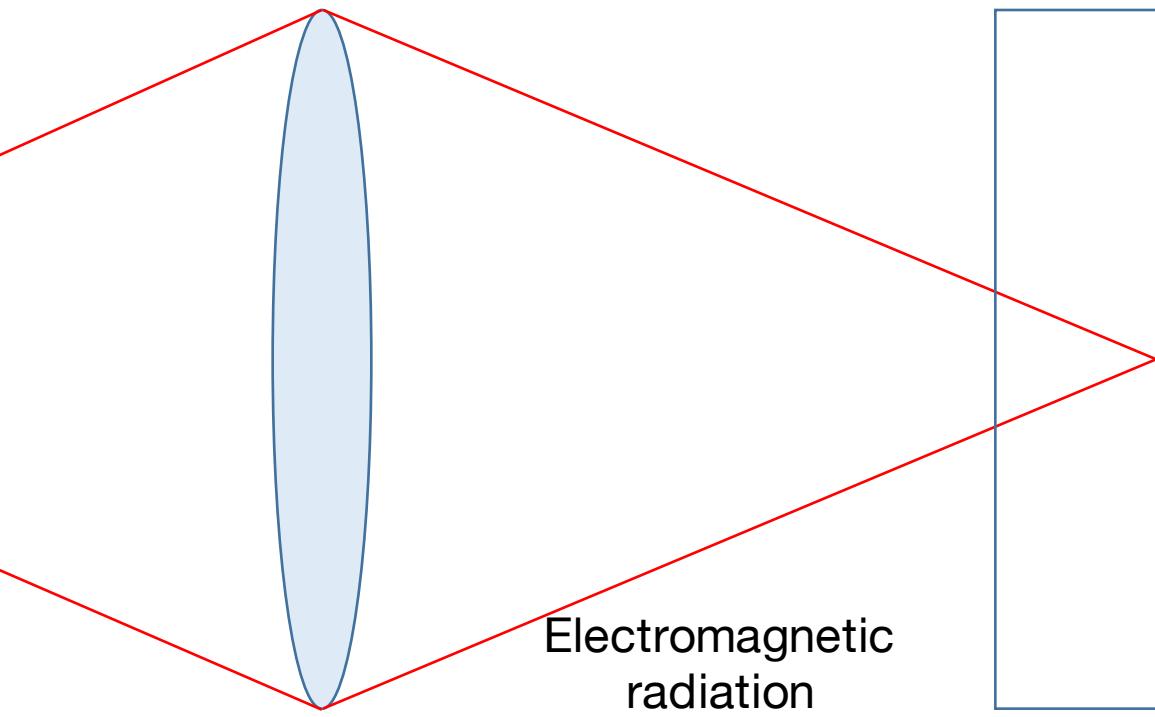
Lecture 2: Mathematical preliminaries for continuous functions

Last time: what is an image?

2. “Physical” Interpretation



Image plane



“Collection”
Element

Electromagnetic
radiation

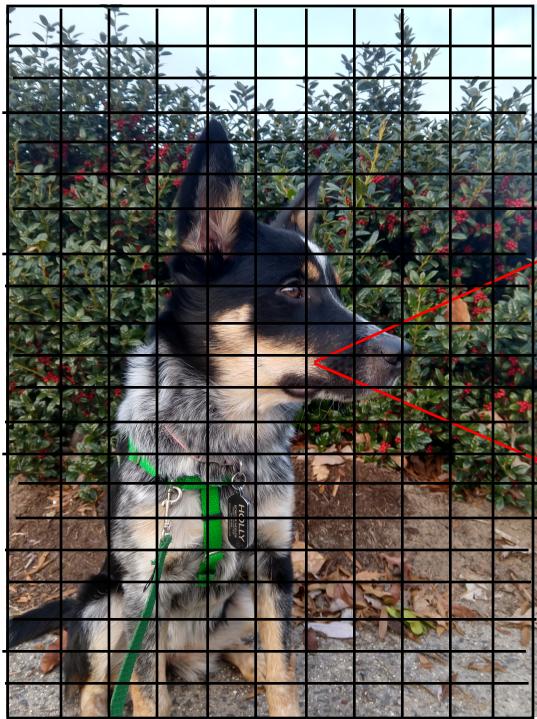
Physical world
(Object plane)

Continuous signal:

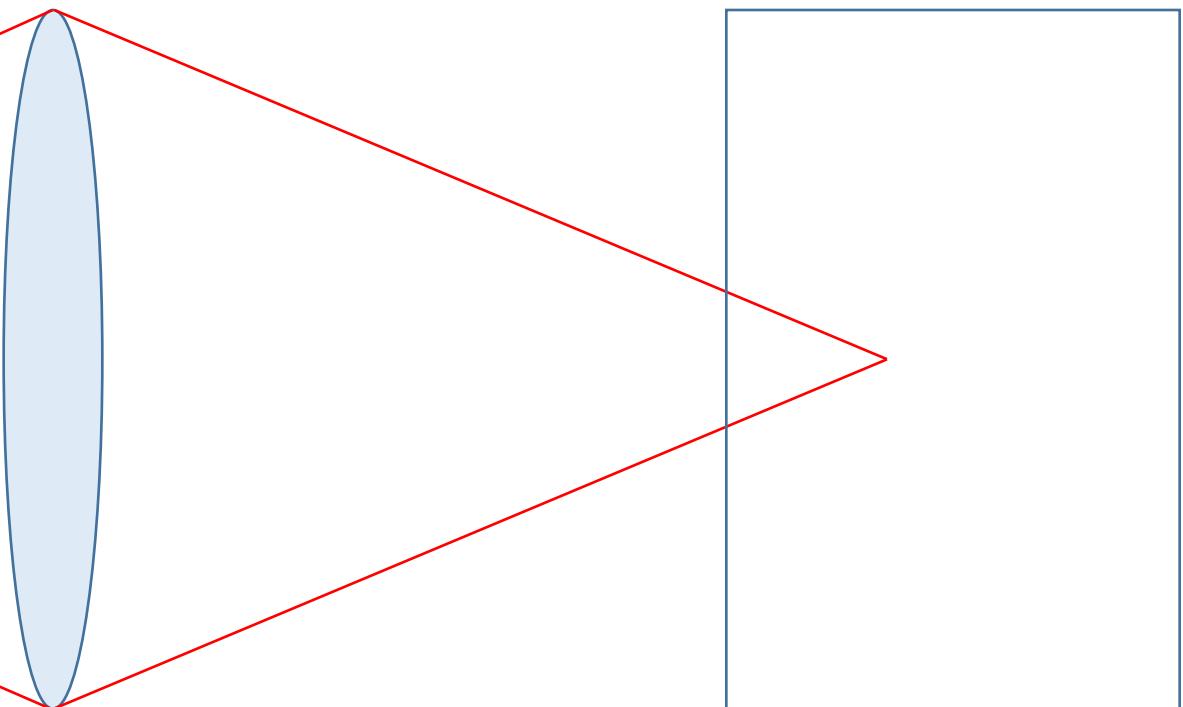
$$I(x, y), (x, y) \in (-\infty, \infty)$$

Last time: what is an image?

$n \times m$ array



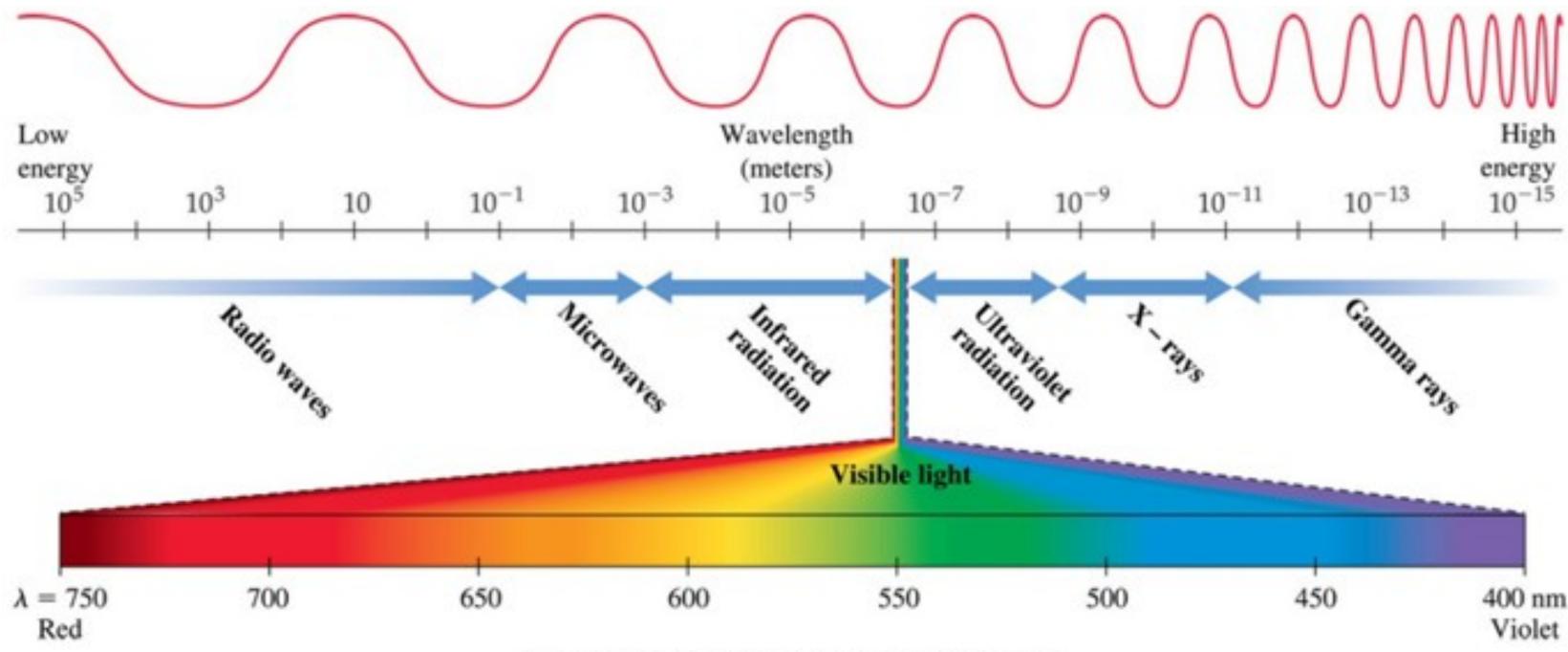
3. “Digital” Interpretation



Photons to electrons → Digitization → Discrete signal

$$I_s(x, y), (x, y) \in Z^{n \times m}$$

Start at the beginning: Electromagnetic waves



Maxwell's equations

$$\nabla \times \vec{\mathcal{E}} = -\mu \frac{\partial \vec{\mathcal{H}}}{\partial t}$$
$$\nabla \times \vec{\mathcal{H}} = \epsilon \frac{\partial \vec{\mathcal{E}}}{\partial t}$$
$$\nabla \cdot \epsilon \vec{\mathcal{E}} = 0$$
$$\nabla \cdot \mu \vec{\mathcal{H}} = 0.$$

Free-space propagation



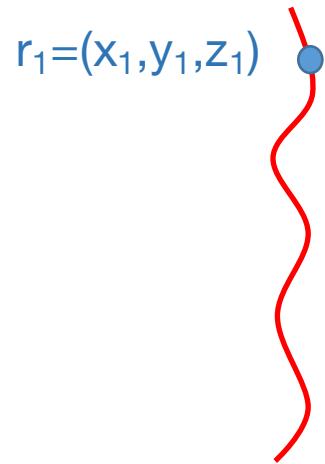
$$\nabla^2 \vec{\mathcal{E}} - \frac{n^2}{c^2} \frac{\partial^2 \vec{\mathcal{E}}}{\partial t^2} = 0$$



$$A(\mathbf{r}_1) \cos(\mathbf{k}\mathbf{r}_1 - \omega t)$$

Scalar solution, 1 freq.

Start at the beginning: optical fields and the black box



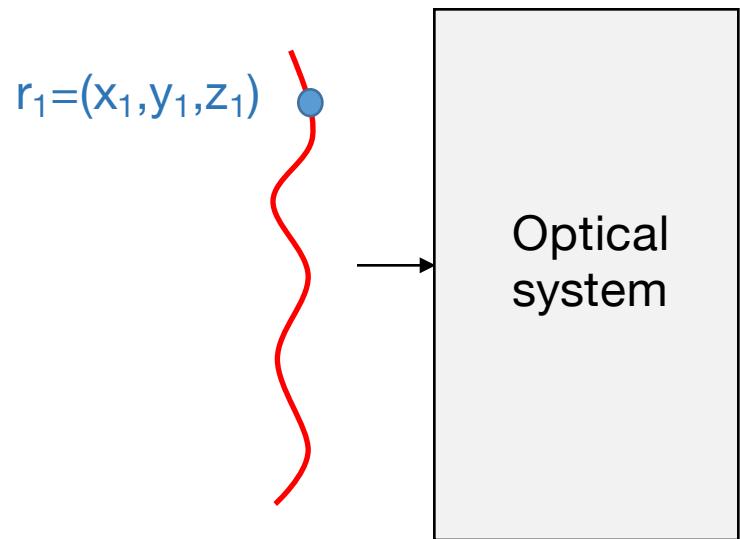
$$U(r_1) = A(r_1) \cos(kr_1 - \omega t)$$

(We will get into the details of optical fields in a few weeks)

The general idea:

1. We will treat light as a wave (an “optical field”)

Start at the beginning: optical fields and the black box

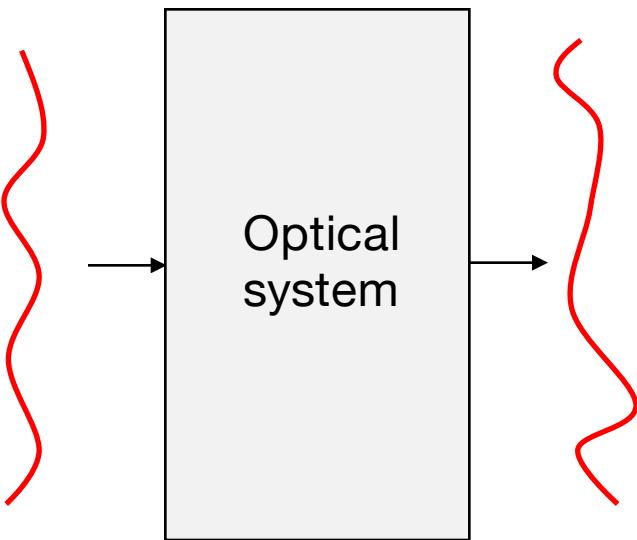


$$U(\mathbf{r}_1) = A(\mathbf{r}_1) \cos(k\mathbf{r}_1 - \omega t)$$

The general idea:

1. We will treat light as a wave (an “optical field”)
2. It enters an optical system, which we treat as a black box
3. This black box has a number of useful properties

Start at the beginning: optical fields and the black box



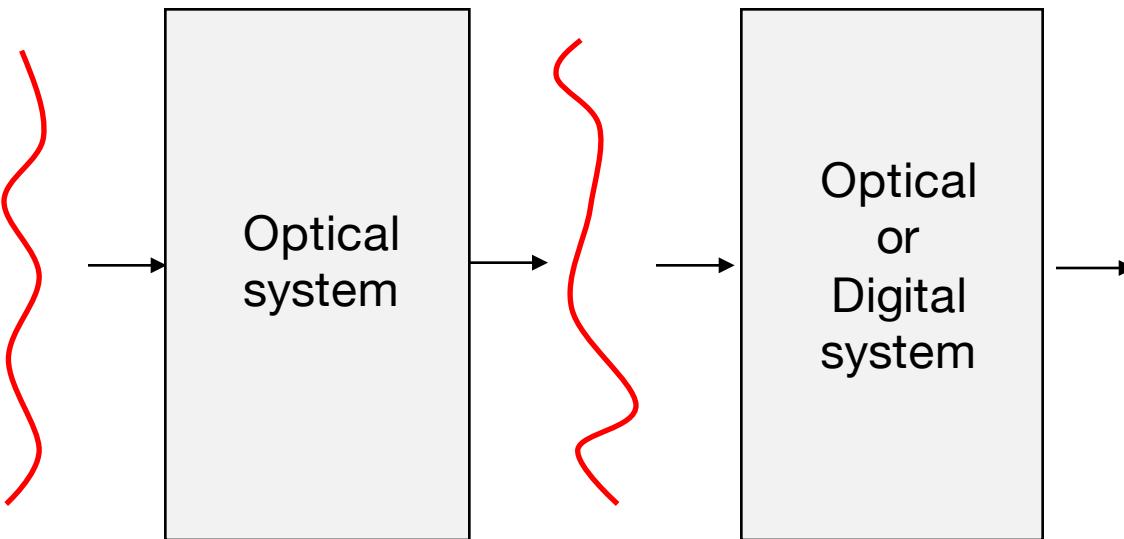
$$A(\mathbf{r}_1) \cos(\mathbf{k}\mathbf{r}_1 - \omega t)$$

$$A(\mathbf{r}_2) \cos(\mathbf{k}\mathbf{r}_2 - \omega t)$$

The general idea:

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2. It enters an optical system, which we treat as a black box
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4. The black box outputs an optical field

Start at the beginning: optical fields and the black box



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The general idea:

1. We will treat light as a wave (an “optical field”)
2. It enters an optical system, which we treat as a black box
3. This black box has a number of useful properties
4. The black box outputs an optical field, which then enters another optical system or a digital system
5. We can cascade these boxes...

Linear systems and the black box

Simplification #1: Let's forget about light changing as a function of time. It does so way too fast, and way too slow:

$$A(r) \cos(kr - \omega t) \rightarrow A(r) \cos(kr)$$

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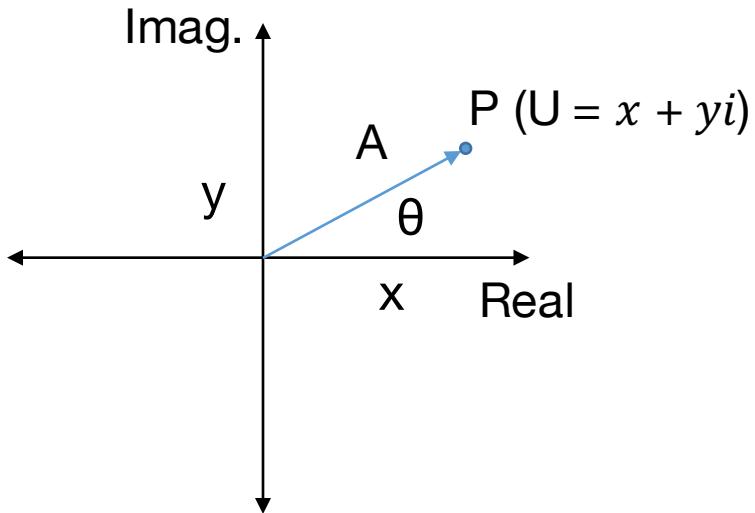
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Simplification #2: We'll use complex numbers when required, it'll make our lives easier. This leads to the *complex field*, $U(r)$:

$$A(r) \cos(kr) \leftrightarrow A(r) e^{ik \cdot r} = U(r)$$

Some things you need to recall about complex numbers

$$U = x + iy, i = \sqrt{-1}$$



$$A = \sqrt{x^2 + y^2}$$

$$\theta = \text{atan}(y/x)$$

More useful representation:

$$x = A \cos\theta$$

$$y = A \sin\theta$$

$$U = A (\cos\theta + i \sin\theta)$$

$$U = A e^{i\theta}$$

A = Amplitude of field

θ = Phase of field

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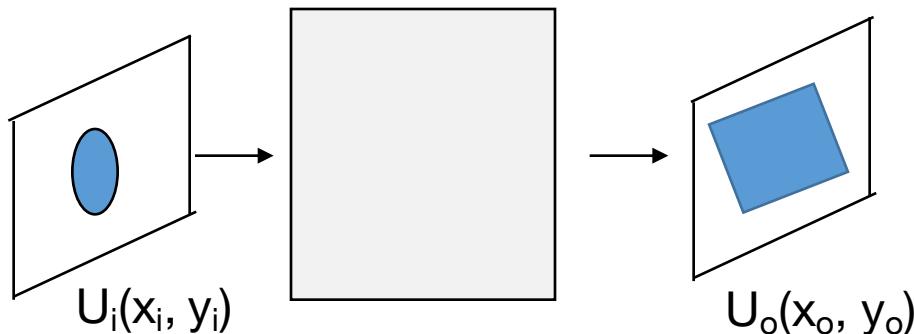
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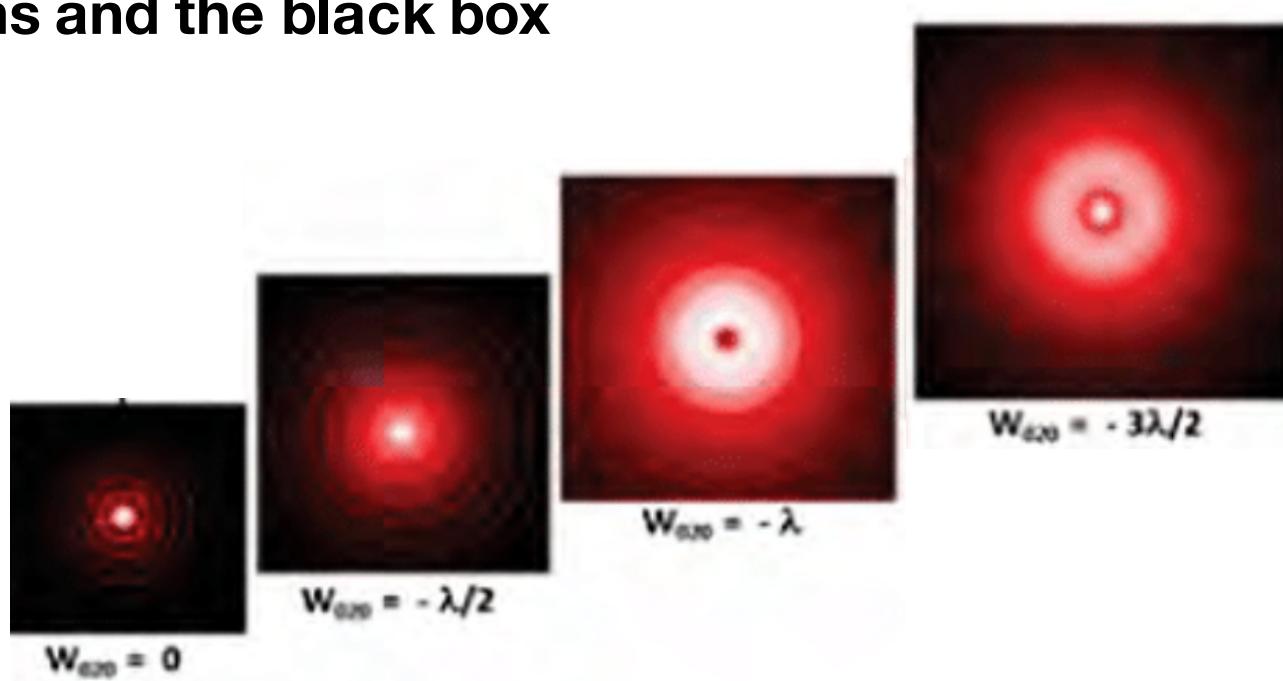
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Simplification #3: Just consider mappings between planes across space. This is a critically important way of thinking for optics. Think “index card 1 to index card 2”.

$$U(r) \rightarrow U(x, y)$$



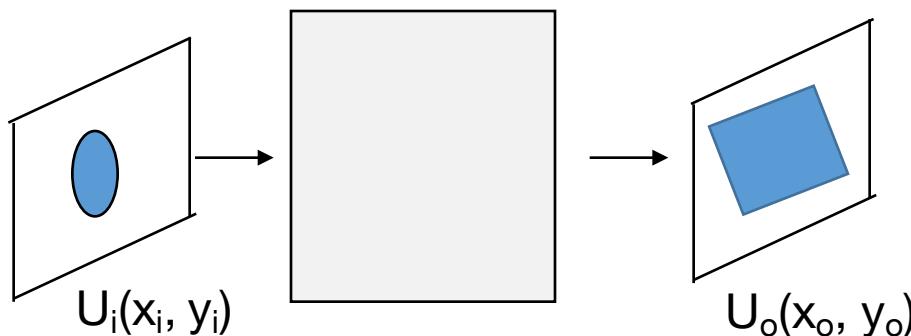
Linear systems and the black box



Propagation of monochromatic light

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$$U(\mathbf{r}) \rightarrow U(x, y)$$



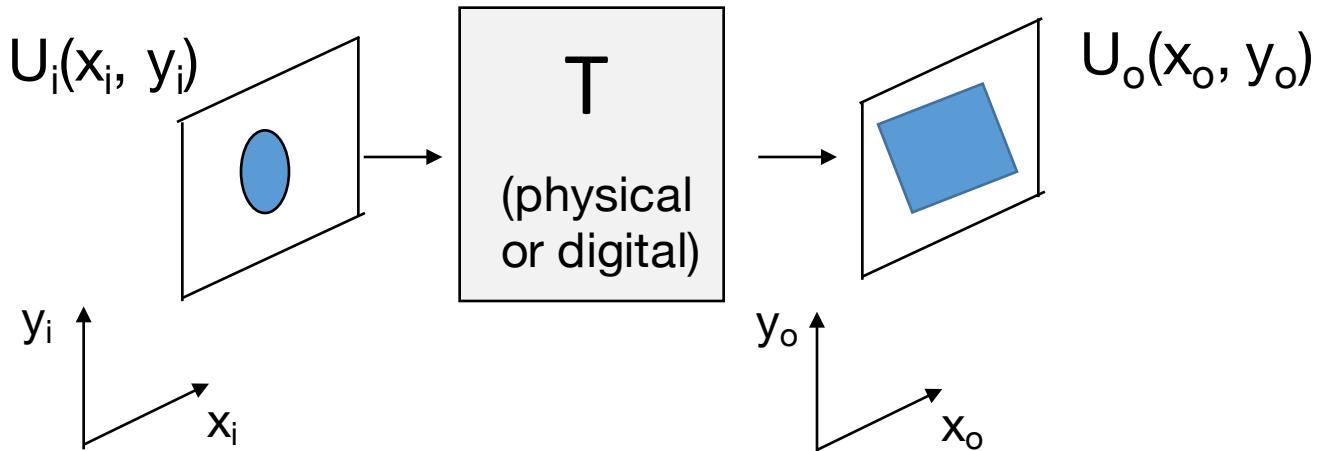
Linear systems and the black box

The “optical” black box system:

An optical black box system maps an input function $U_i(x_i, y_i)$ to an output function $U_o(x_o, y_o)$ via a transform T :

$$U_o(x_o, y_o) = T [U_i(x_i, y_i)]$$

Where $T[]$ denotes the optical black box transformation



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Important properties of linear systems:

1. Homogeneity and additivity (superposition):

$$T [aU_1(x, y) + bU_2(x, y)] = aT [U_1(x, y)] + bT [U_2(x, y)]$$

2. Shift invariance: for shift distances d_x and d_y , we assume that,

$$U_o(x_o - d_x, y_o - d_y) = T [U_i(x_i - d_x, y_i - d_y)]$$

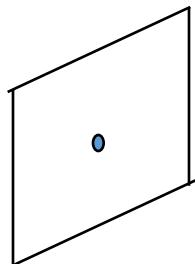
Black box transforms as a convolution

Assuming 1) linearity and 2) shift-invariance, we can model any black box with 1 piece of information:

Input Dirac delta function into the black box:

$$\delta(x) = \begin{cases} +\infty, & x = 0 \\ 0, & x \neq 0 \end{cases}$$

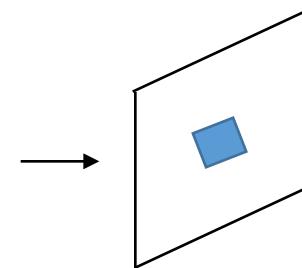
A “perfect”
point
source



$$\delta(x_i, y_i)$$



LSI
system



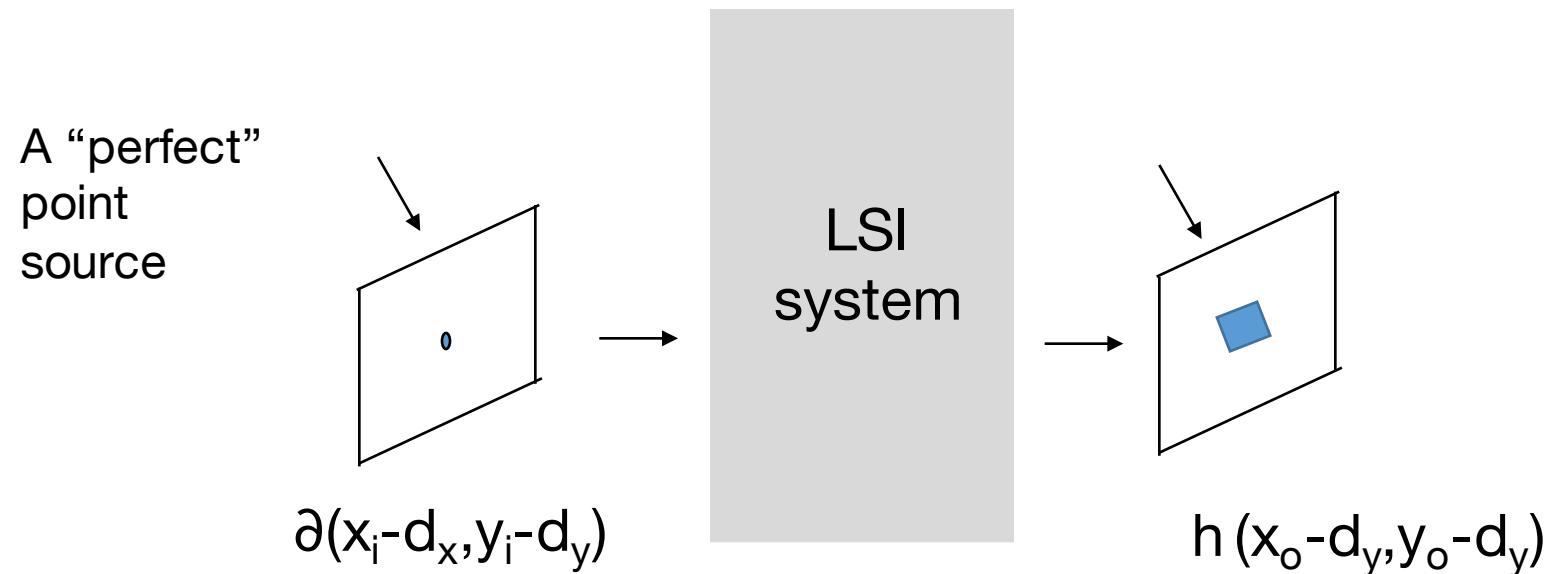
$$h(x_o, y_o)$$

$$h(x_o, y_o) = T [\delta(x_i, y_i)]$$

Black box transforms as a convolution

Assuming 1) linearity and 2) shift-invariance, we can model any black box with 1 piece of information:

We know the system is shift invariant:



$$h(x_o - d_y, y_o - d_y) = T [\partial(x_i - d_x, y_i - d_y)]$$

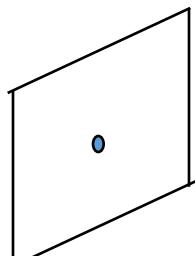
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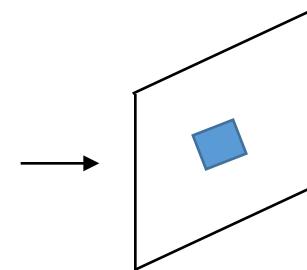
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$$\delta(x_i, y_i)$$

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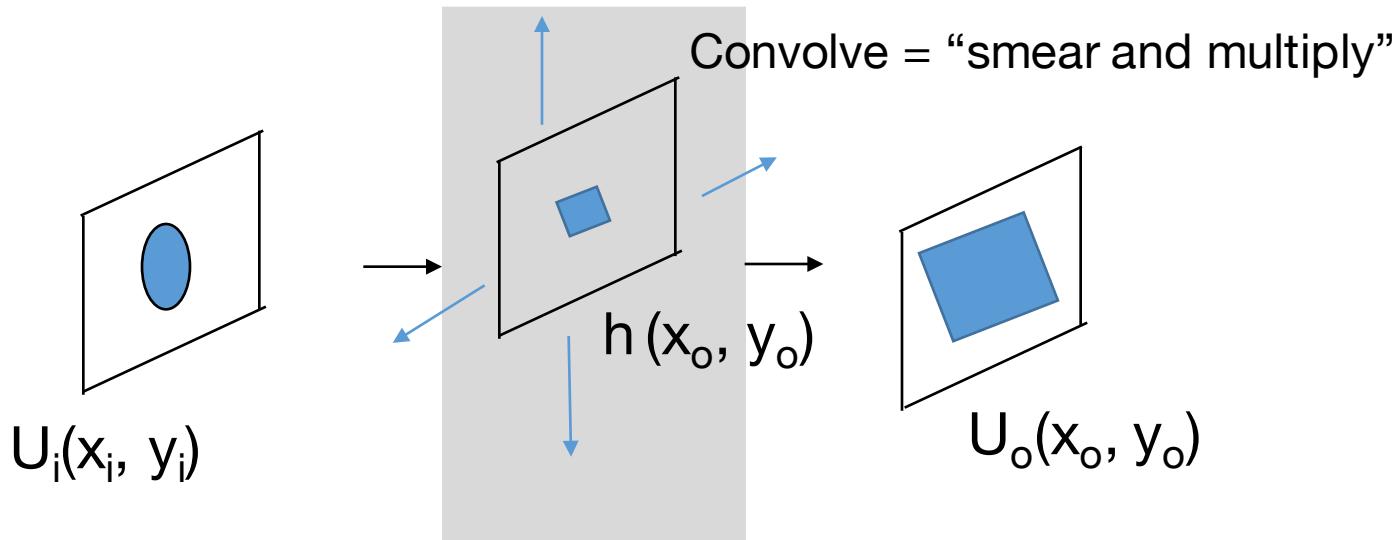
$h(x_o, y_o)$ is the
system’s point-
spread function

Point-spread function

$$h(x_o, y_o) = T [\delta(x_i, y_i)]$$

Black box transforms as a convolution

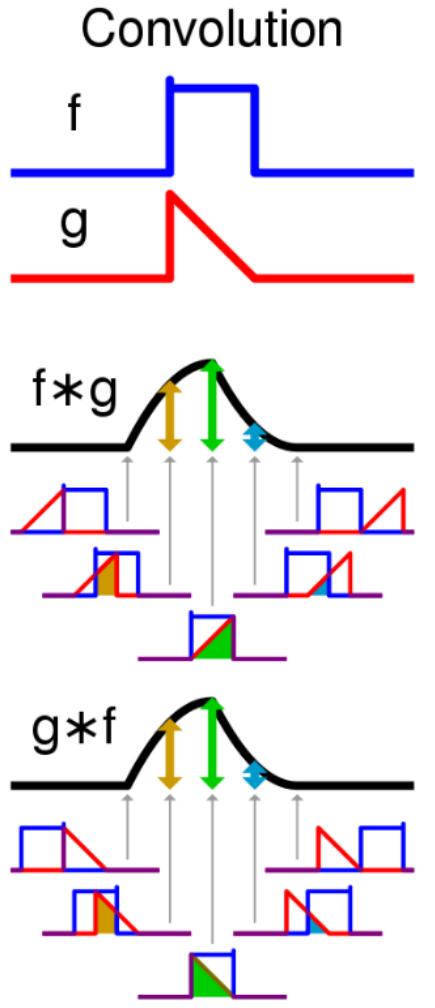
Knowing the point-spread function, it is direct to model any output of the black box, given an input:



$$U_o(x_o, y_o) = \iint_{-\infty}^{\infty} U_i(x_i, y_i) h(x_o - x_i, y_o - y_i) dx_i dy_i$$

Output of linear system is a convolution of the input with its point-spread function

1D convolution example



Steps to perform a convolution:

1. Flip one signal (the second one = the PSF)
2. Position PSF right before overlap
3. Step PSF over to position x_o
4. Compute *area* of overlap of two functions
5. Convolution value at x_o = area of overlap
6. Repeat 3-5 until signals do not overlap

With incremental steps:

2D convolution example

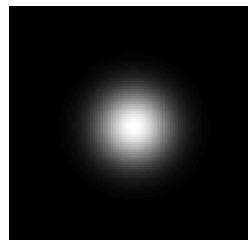
- Direct extension of 1D concept to 2D functions
 - Note – it is effectively the same with discrete functions = matrices

$$U_1(x,y)$$



*

y2



x2

Y

—

$$U_0(x,y)$$



X

Useful properties of the convolution

1. Commutativity $U(x) * h(x) = h(x) * U(x)$

⇒ You can choose which signal to “flip”

2. Associativity $U(x) * [V(x) * W(x)] = [U(x) * V(x)] * W(x)$

⇒ Can change order → sometimes one order is easier than another

3. Distributivity $U(x) * [h_1(x) * h_2(x)] = U(x) * h_1(x) + U(x) * h_2(x)$

Signals in space and spatial frequency

- What we have so far:
 - Continuous & (possibly) complex function for images across space
 - Black-box linear transformation from one domain to the next via convolution
- Analogy:
 - Time-varying voltage/current going through a circuit
 - Audio signal passing through a filter

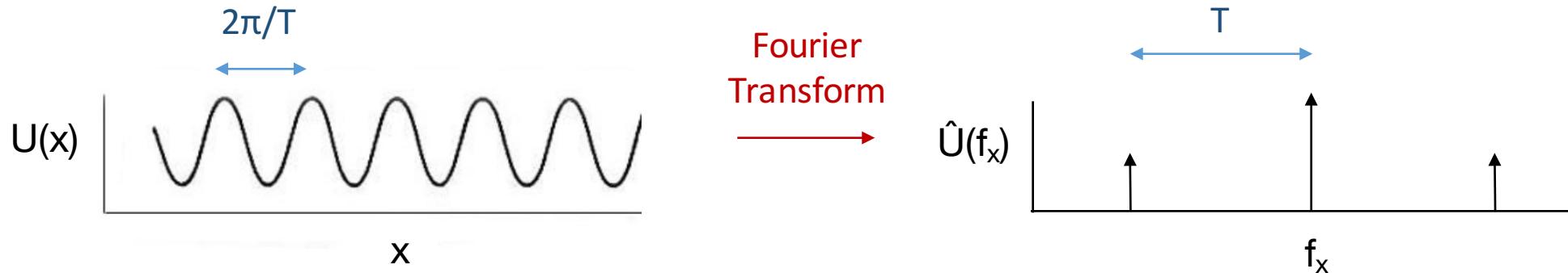


Complex function of time -> frequency

Fourier Transforms

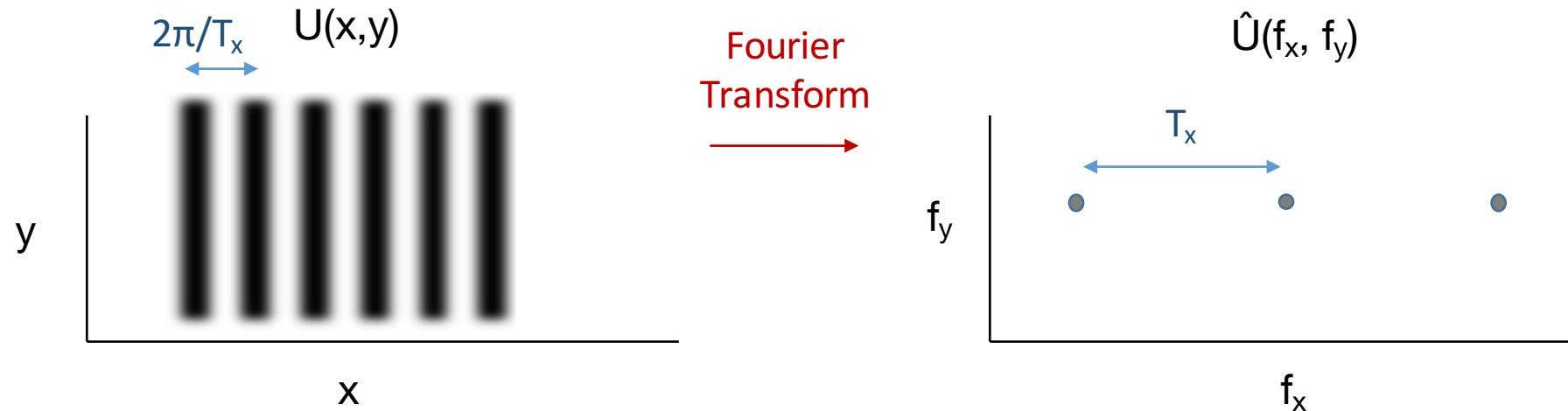
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 - Here, we have 2D (complex) function across space (x, y) -> *spatial frequency* (f_x, f_y)
- $\left. \right\}$ Complex function of time -> frequency



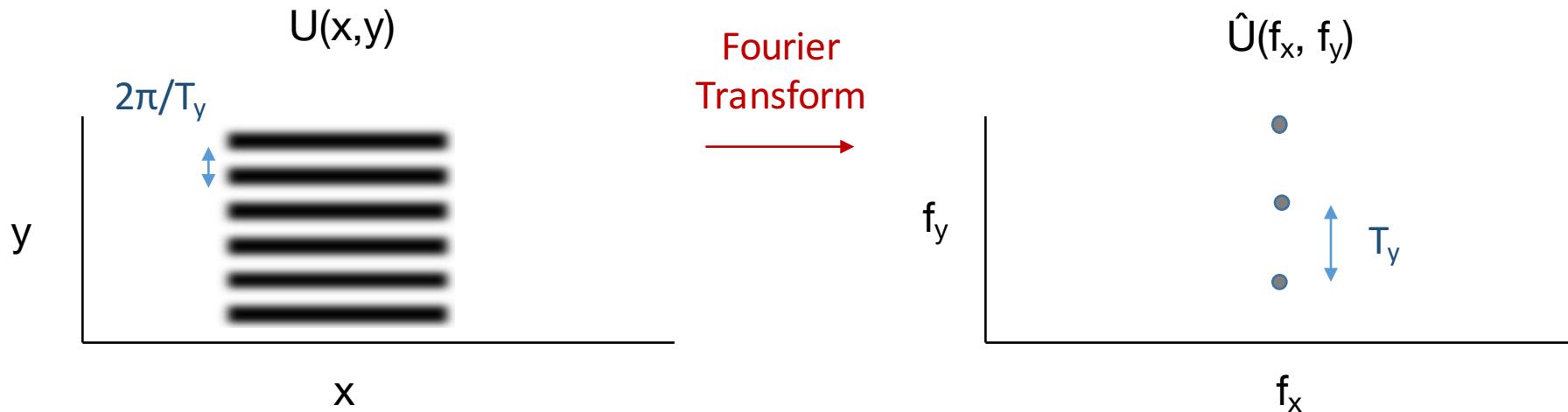
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Continuous Fourier transforms – for 2D images

Decomposition of a signal into elementary functions of form, $\exp(-2\pi i(f_x x + f_y y))$:

$$\mathcal{F}\{U(x, y)\} = \hat{U}(f_x, f_y) = \iint_{-\infty}^{\infty} U(x, y) \exp(-2\pi i(f_x x + f_y y)) dx dy$$

U is absolutely integrable & no infinite discontinuities. The inverse Fourier transform is,

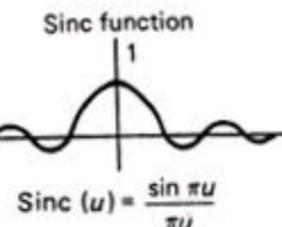
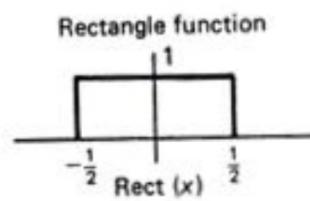
$$\mathcal{F}^{-1}\{\hat{U}(f_x, f_y)\} = U(x, y) = \iint_{-\infty}^{\infty} \hat{U}(f_x, f_y) \exp(2\pi i(f_x x + f_y y)) df_x df_y$$

Additional Details:

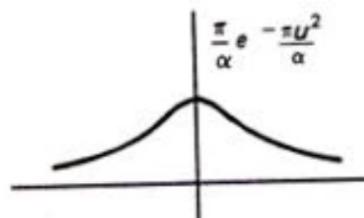
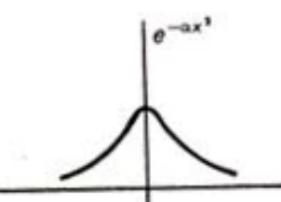
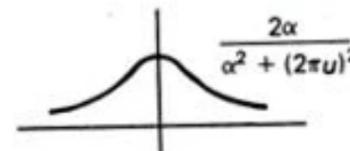
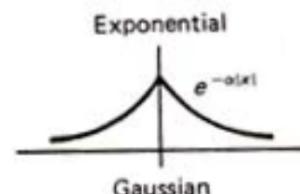
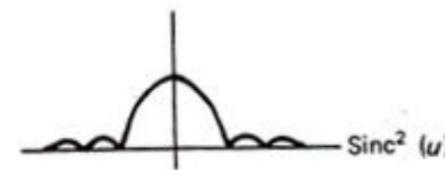
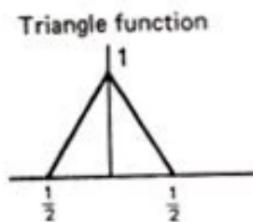
- Goodman Chapter 2.1
- Mathworld/Wikipedia, Fourier Transform

A few examples of Fourier transform pairs, 1D

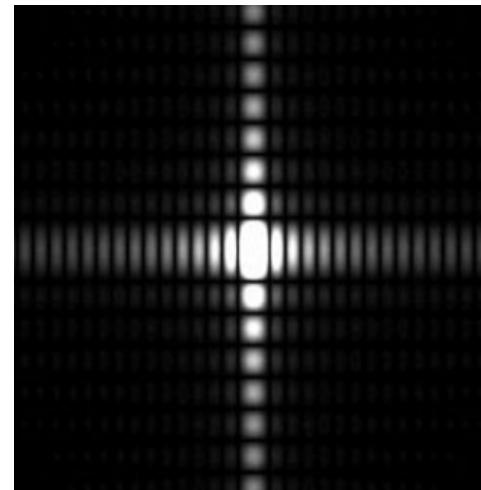
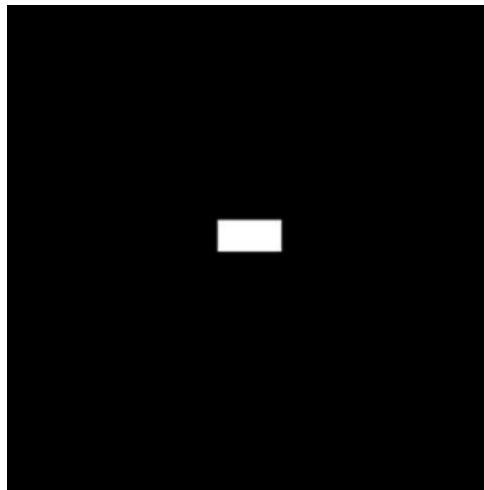
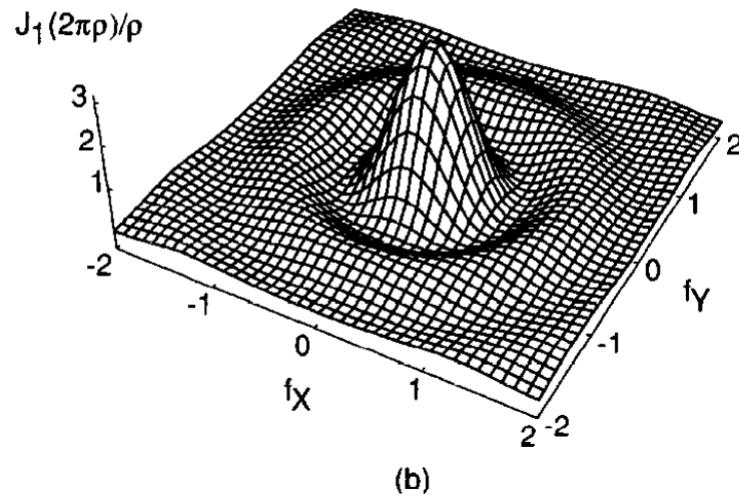
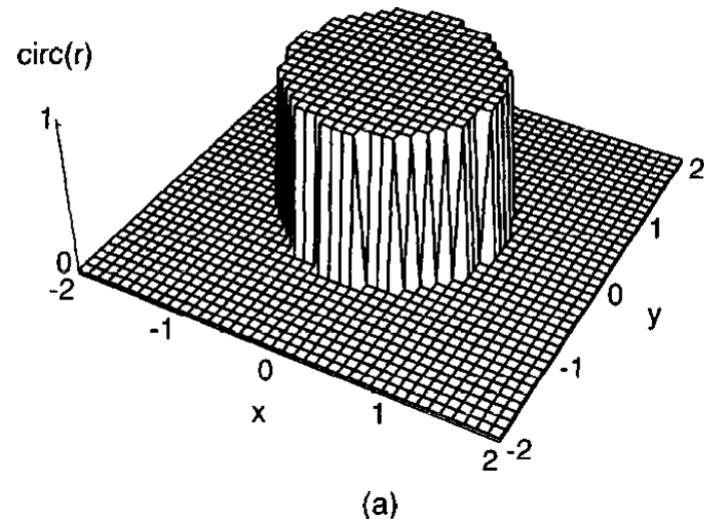
$U(x)$



$\hat{U}(f_x)$



Examples of Fourier transform pairs, 2D



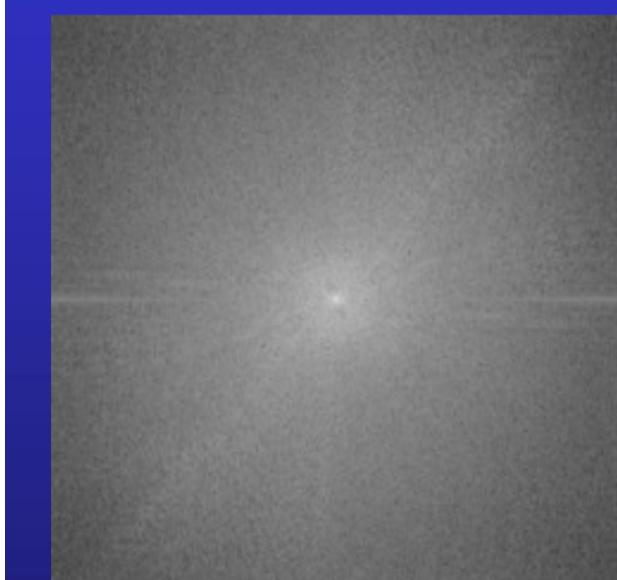
$U_1(x,y)$ $U_2(x,y)$ 

Cheetah

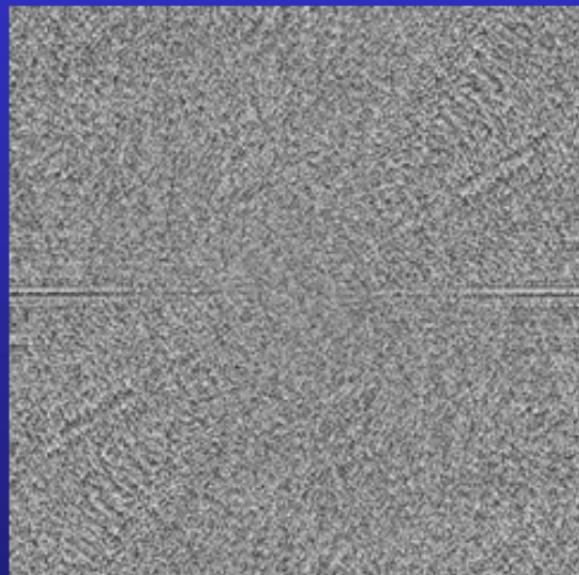


Zebra

$$\hat{U}_1(f_x, f_y)$$

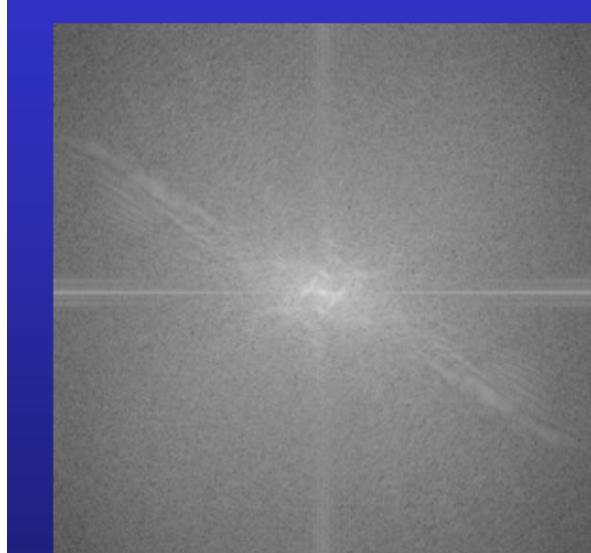


magnitude of cheetah

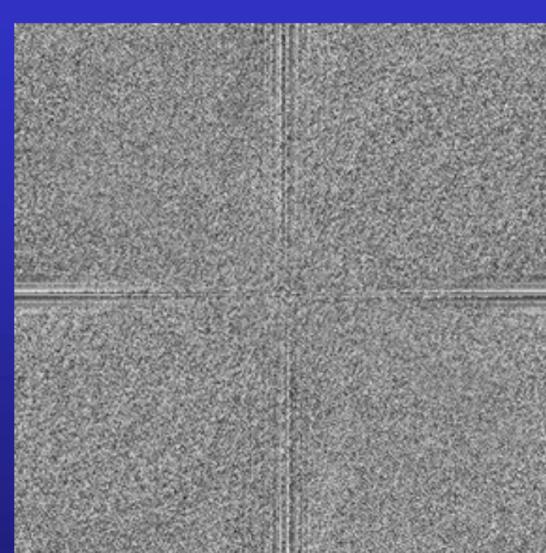


phase of cheetah

$$\hat{U}_2(f_x, f_y)$$



magnitude of zebra



phase of zebra

Important properties of the Fourier transform

- Linearity
- Scaling
- Shift
- Parseval's Theorem (energy conservation)
- Fourier integral theorem

Additional Details:

- Goodman Chapter 2.1
- Mathworld/Wikipedia, Fourier Transform

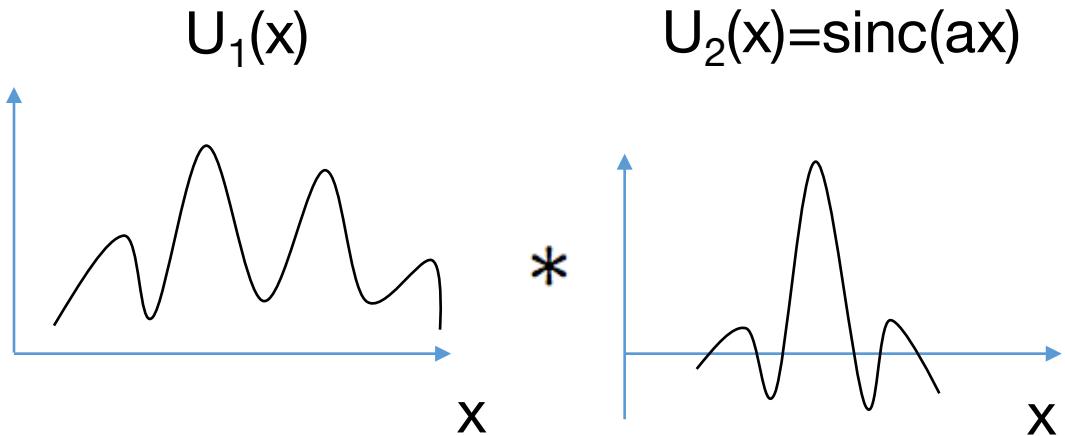
Convolution - Fourier Transform relationship: Convolution Theorem

Convolution theorem. If $\mathcal{F}\{g(x, y)\} = G(f_X, f_Y)$ and $\mathcal{F}\{h(x, y)\} = H(f_X, f_Y)$, then

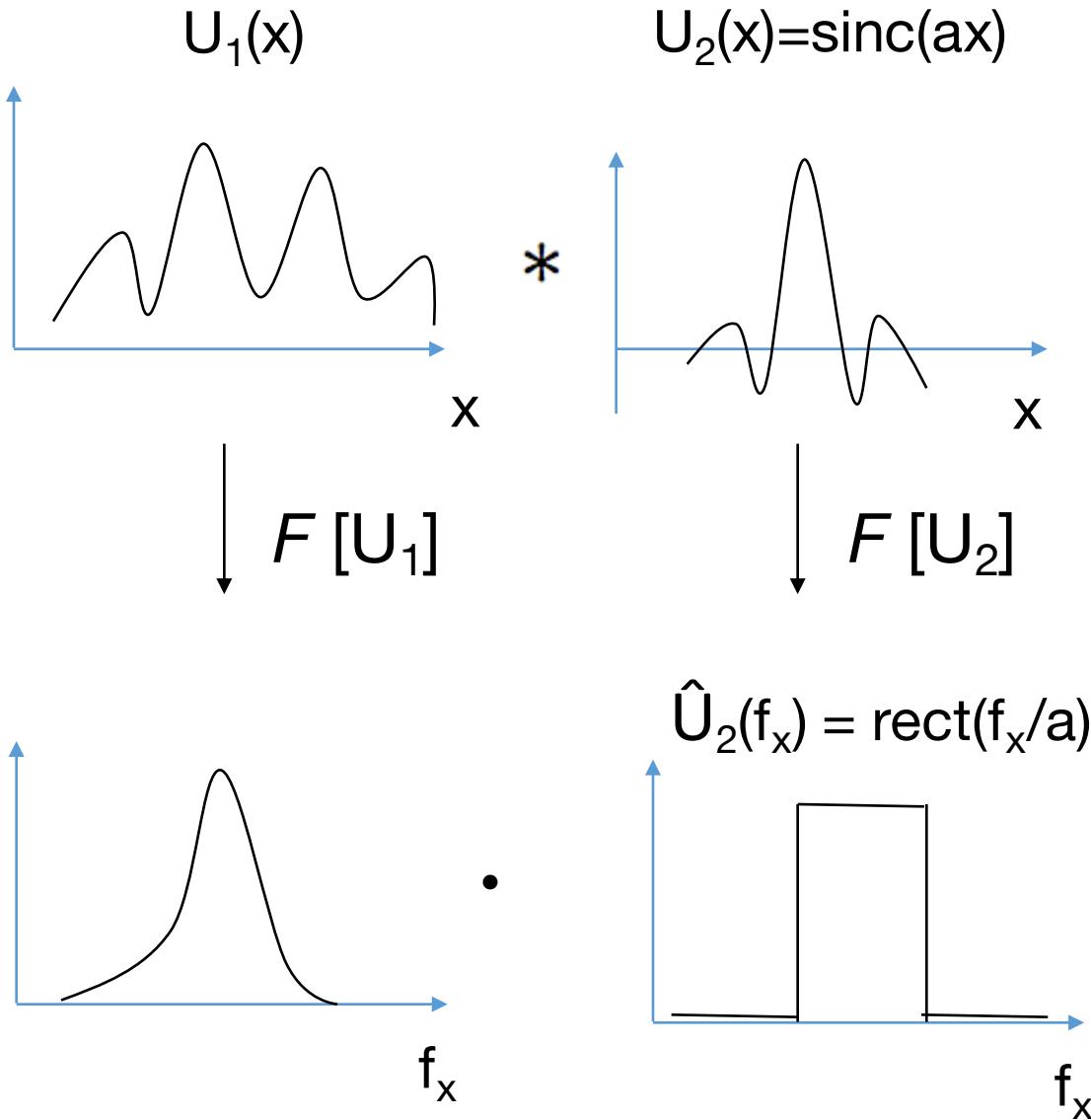
$$\mathcal{F} \left\{ \iint_{-\infty}^{\infty} g(\xi, \eta) h(x - \xi, y - \eta) d\xi d\eta \right\} = G(f_X, f_Y) H(f_X, f_Y).$$

“The convolution of two functions in space can be performed by a multiplication in the Fourier domain (spatial frequency domain)”

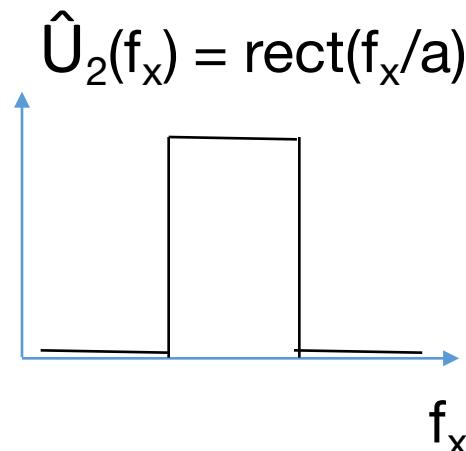
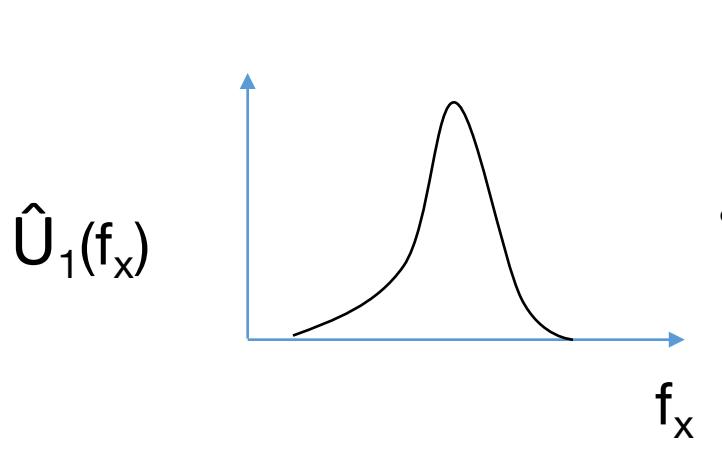
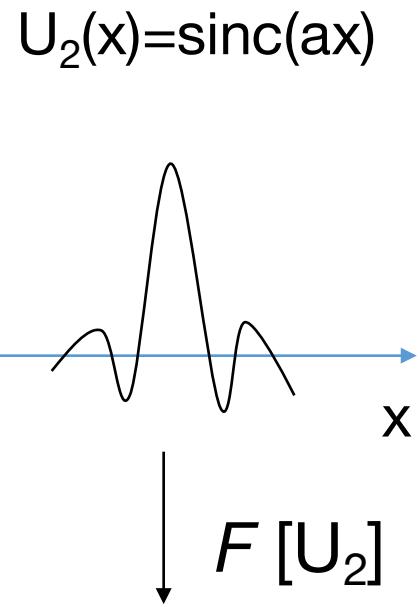
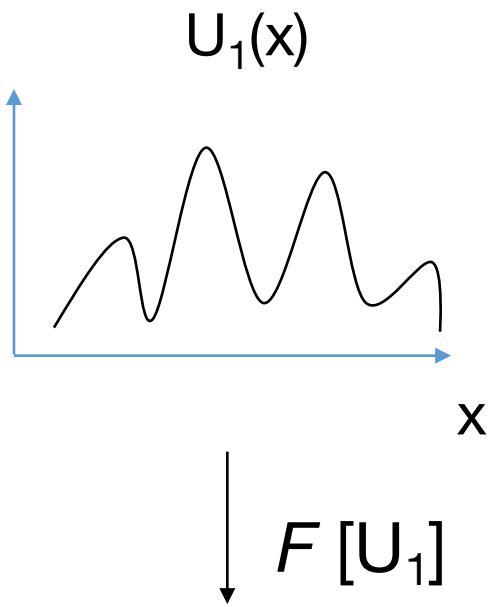
Example of convolution theorem, 1D



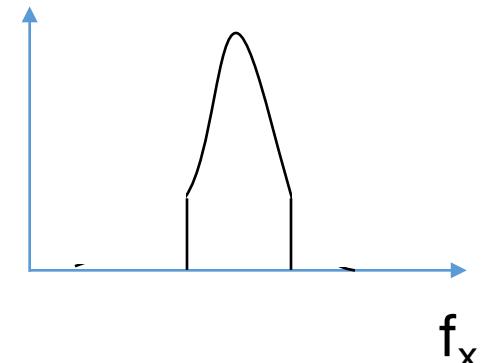
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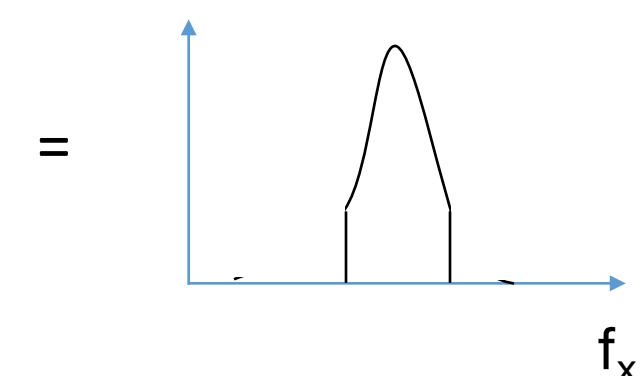
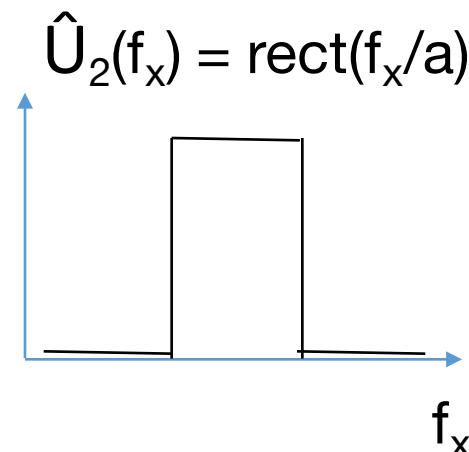
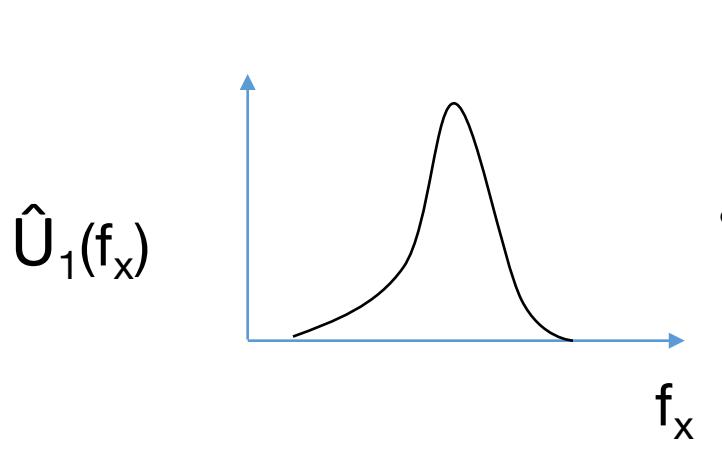
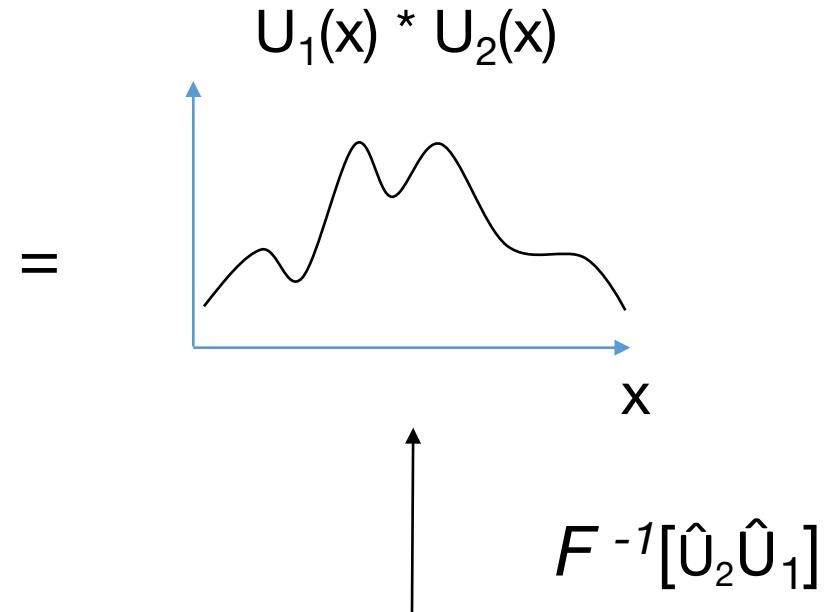
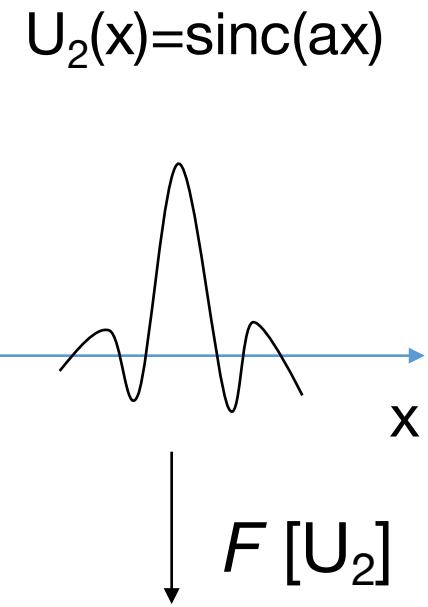
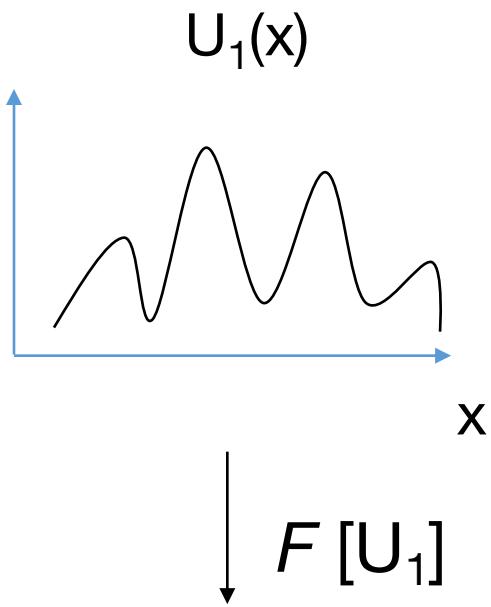
Example of convolution theorem, 1D



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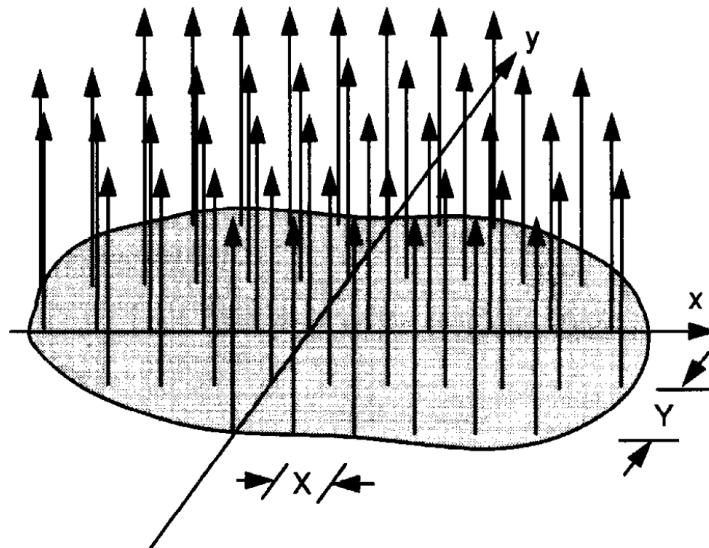


Example of convolution theorem, 1D



The Sampling Theorem – from Goodman Section 2.4.1

$$U_s(x, y) = \text{comb}(x/X)\text{comb}(y/Y)U(x, y)$$



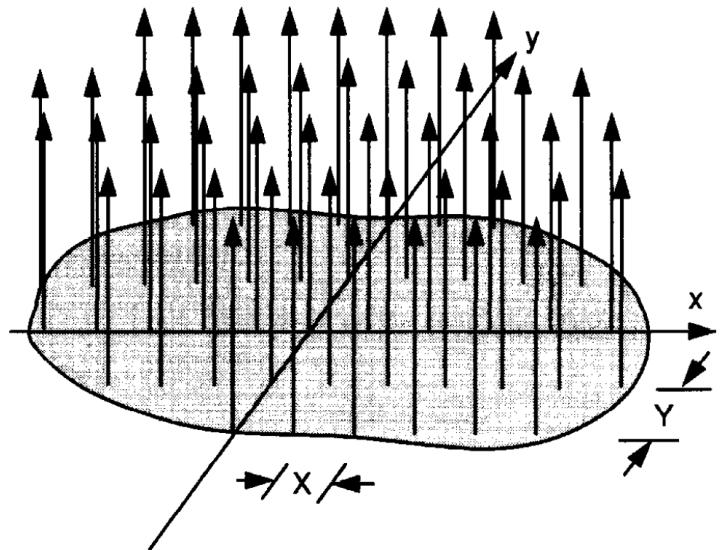
Signal sampling occurs with:

- CMOS (pixel) sensors, PMTs, SPADs
- A-to-D after antennas
- A-to-D after acoustic transducers

Sampling interval width X and Y

The Sampling Theorem – from Goodman Section 2.4.1

$$U_s(x, y) = \text{comb}(x/X)\text{comb}(y/Y)U(x, y)$$



Sampling interval width X and Y

$$\hat{U}_s(f_x, f_y) = \mathcal{F} [\text{comb}(x/X)\text{comb}(y/Y)] * \hat{U}(f_x, f_y)$$

The Sampling Theorem – from Goodman Section 2.4.1

$$\hat{U}_s(f_x, f_y) = \mathcal{F} [\text{comb}(x/X)\text{comb}(y/Y)] * \hat{U}(f_x, f_y)$$

The Sampling Theorem – from Goodman Section 2.4.1

$$\hat{U}_s(f_x, f_y) = \mathcal{F} [\text{comb}(x/X)\text{comb}(y/Y)] * \hat{U}(f_x, f_y)$$

$$\mathcal{F} [\text{comb}(x/X)\text{comb}(y/Y)] = \sum_{n=-\infty}^{\infty} \sum_{m=-\infty}^{\infty} \delta \left(f_x - \frac{n}{X}, f_y - \frac{m}{Y} \right)$$

The Sampling Theorem – from Goodman Section 2.4.1

$$\hat{U}_s(f_x, f_y) = \mathcal{F} [\text{comb}(x/X)\text{comb}(y/Y)] * \hat{U}(f_x, f_y)$$

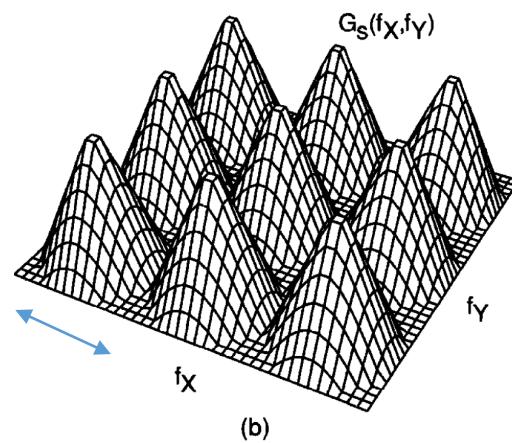
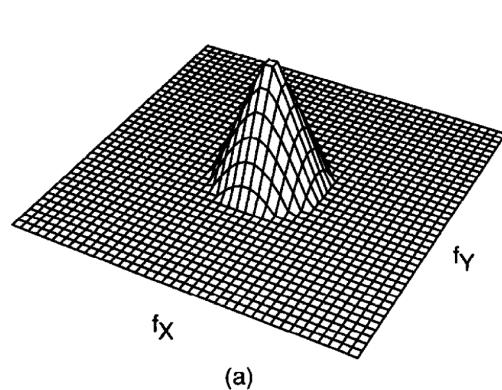
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Signal extends from $(-B_x, -B_y)$
to (B_x, B_y) in Fourier domain

$$\text{rect} \left(\frac{f_x}{2B_x} \right) \text{rect} \left(\frac{f_y}{2B_y} \right) \hat{U}_s(f_x, f_y) = \hat{U}(f_x, f_y)$$

Bandwidth (B_x, B_y) of signal

$$\operatorname{rect}\left(\frac{f_x}{2B_x}\right) \operatorname{rect}\left(\frac{f_y}{2B_y}\right) \hat{U}_s(f_x, f_y) = \hat{U}(f_x, f_y)$$

$$\mathcal{F}[\bullet] \quad h(x, y) = 4B_x B_y \operatorname{sinc}(2B_x x) \operatorname{sinc}(2B_y y)$$

$$\operatorname{rect}\left(\frac{f_x}{2B_x}\right) \operatorname{rect}\left(\frac{f_y}{2B_y}\right) \hat{U}_s(f_x, f_y) = \hat{U}(f_x, f_y)$$

$$\mathcal{F}[\bullet] \quad h(x, y) = 4B_x B_y \operatorname{sinc}(2B_x x) \operatorname{sinc}(2B_y y)$$

$$h(x, y) * (U(x, y) \operatorname{comb}(x/X) \operatorname{comb}(y/Y)) = U(x, y)$$

$$\text{rect}\left(\frac{f_x}{2B_x}\right) \text{rect}\left(\frac{f_y}{2B_y}\right) \hat{U}_s(f_x, f_y) = \hat{U}(f_x, f_y)$$

$$\mathcal{F}[\bullet] \quad h(x, y) = 4B_x B_y \text{sinc}(2B_x x) \text{sinc}(2B_y y)$$

$$h(x, y) * (U(x, y) \text{comb}(x/X) \text{comb}(y/Y)) = U(x, y)$$

$$U(x, y) \text{comb}(x/X) \text{comb}(y/Y) = XY \sum_{n=-\infty}^{\infty} \sum_{m=-\infty}^{\infty} U(nX, mY) \delta(x - nX, y - mY)$$

$$\text{rect}\left(\frac{f_x}{2B_x}\right) \text{rect}\left(\frac{f_y}{2B_y}\right) \hat{U}_s(f_x, f_y) = \hat{U}(f_x, f_y)$$


 $F[\bullet]$ $h(x, y) = 4B_x B_y \text{sinc}(2B_x x) \text{sinc}(2B_y y)$

$$h(x, y) * (U(x, y) \text{comb}(x/X) \text{comb}(y/Y)) = U(x, y)$$

$$U(x, y) \text{comb}(x/X) \text{comb}(y/Y) = XY \sum_{n=-\infty}^{\infty} \sum_{m=-\infty}^{\infty} U(nX, mY) \delta(x - nX, y - mY)$$

$$U(x, y) = 4B_x B_y XY \sum_{n=-\infty}^{\infty} \sum_{m=-\infty}^{\infty} U(nX, mY) \text{sinc}[2B_x(x - nX)] \text{sinc}[2B_y(y - mY)]$$

The Sampling Theorem

When sampled appropriately, a discrete signal can *exactly* reproduce a continuous signal:

$$\underline{U(x,y)} = 4B_x B_y XY \sum_{n=-\infty}^{\infty} \sum_{m=-\infty}^{\infty} \underline{U(nX, mY) \text{sinc}[2B_x(x - nX)] \text{sinc}[2B_y(y - mY)]}$$

Continuous signal:

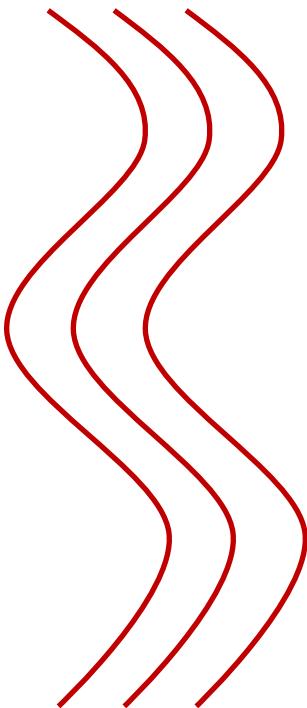
- EM field
- Sound wave
- MR signal

Discretized signal:

- Detected EM field
- Sampled sound wave
- Sampled MR signal

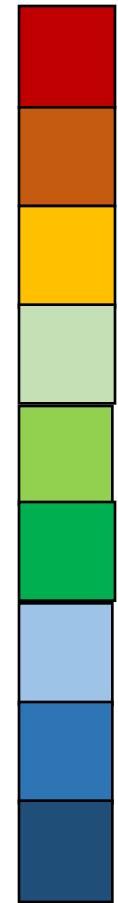
What does the Sampling Theorem mean for us?

Continuous fields



*conditions

Discretize vectors
(and matrices)



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