## Visual Computing: Fourier Transform <br> Prof. Marc Pollefeys

## Last week

## Correlation

(e.g. Template-matching)


## Convolution

(e.g. point spread function)

$$
I^{\prime}=\sum_{j=-k}^{k} \sum_{i=-k}^{k} K(i, j) I(x+i, y+j)
$$

(matlab default)


$$
I^{\prime}=\sum_{j=-k}^{k} \sum_{i=-k}^{k} K(i, j) I(x-i, y-j)
$$

## Last lecture

- Edges
- Maximal gradients
- Zero-crossing Laplacian


Canny edge detection

- Hough transform
- Corners
- Maximal difference with neighbors



## Scale Invariant Feature Transform (SIFT)



## ЕНН

# Visual Computing: Fourier Transform <br> Prof. Marc Pollefeys 

## Aliasing

- Can' t shrink an image by taking every second pixel
- If we do, characteristic errors appear
- In the next few slides
- Typically, small phenomena look bigger; fast phenomena can look slower
- Common phenomenon
- Wagon wheels rolling the wrong way in movies
- Checkerboards misrepresented in ray tracing
- Striped shirts look funny on color television



Constructing a pyramid by taking every second pixel leads to layers that badly misrepresent the top layer


## Open questions

- What causes the tendency of differentiation to emphasize noise?
- In what precise respects are discrete images different from continuous images?
- How do we avoid aliasing?
- General thread: a language for fast changes

The Fourier Transform

## The Fourier Transform

- Represent function on a new basis
- Think of functions as vectors, with many components
- We now apply a linear transformation to transform the basis
- dot product with each basis element
- In the expression, $u$ and $v$ select the basis element, so a function of $x$ and $y$ becomes a function of $u$ and $v$
- basis elements have the form $e^{-i 2 \pi(u x+v y)}$

$$
=\cos 2 \pi(u x+v y)-i \sin 2 \pi(u x+v y)
$$

$$
F(g(x, y))(u, v)=\iint_{\mathrm{R}^{2}} g(x, y) e^{-i 2 \pi(u x+v y)} d x d y
$$

Discrete FT: transformed image $\longrightarrow F=\mathbf{U} f$ vectorized image Fourier transform base,
also possible Wavelets, steerable pyramids, etc.

## Fourier transform and linear systems

- Basis functions of Fourier transform are eigenfunctions of linear systems
(or why Electrical Engineers love the Fourier transform)


Fourier basis element

$$
e^{-i 2 \pi(u x+v y)}
$$

example, real part
$\mathrm{F}^{\mathrm{u}, \mathrm{v}}(\mathrm{x}, \mathrm{y})$
$\mathrm{Fu}, \mathrm{v}(\mathrm{x}, \mathrm{y})=$ const. for (ux+vy)=const.

Vector (u,v)

- Magnitude gives frequency
- Direction gives orientation.


Here $u$ and $v$ are larger than in the previous slide.


## And larger still



## Fourier basis functions

$$
\begin{aligned}
& e^{-i 2 \pi(u x+v y)} \\
& =\cos 2 \pi(u x+v y)-i \sin 2 \pi(u x+v y)
\end{aligned}
$$

(note: basis functions are global)


## ETH



## Phase and Magnitude

- Fourier transform of a real function is complex
- difficult to plot, visualize
- instead, we can think of the phase and magnitude of the transform
- Phase is the phase of the complex transform
- Magnitude is the magnitude of the complex transform
- Curious fact
- all natural images have about the same magnitude transform
- hence, phase seems to matter, but magnitude largely doesn' $t$
- Demonstration
- Take two pictures, swap the phase transforms, compute the inverse - what does the result look like?


This is the magnitude transform of the cheetah pic

This is the phase transform of the cheetah pic



This is the magnitude transform of the zebra pic

This is the phase
transform of the zebra pic


## Reconstruction with zebra phase, cheetah magnitude



## Reconstruction

 with cheetah phase, zebra magnitude


## ETH



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## Various Fourier Transform Pairs

- Important facts
- The Fourier transform is linear
- There is an inverse FT $\quad f=\mathbf{U}^{-1} F$
- scale function down $\Leftrightarrow$ scale transform up
i.e. high frequency = small details
- The FT of a Gaussian is a Gaussian.

compare to box function transform


Fourier Transform of important functions


* see in a few slides


## Convolution theorem

- The convolution theorem
- The Fourier transform of the convolution of two functions is the product of their Fourier transforms

$$
F . G=\mathbf{U}(f * * g) \quad \text { (cfr. filtering) }
$$

- The Fourier transform of the product of two functions is the convolution of the Fourier transforms

$$
\begin{equation*}
F * * G=\mathbf{U}(f . g) \tag{cfr.sampling}
\end{equation*}
$$

## EPM

## Sampling

- Go from continuous world to discrete world, from function to vector
- Samples are typically measured on regular grid




## A continuous model for a sampled

## function

- We want to be able to
approximate integrals
sensibly $\quad\left[\right.$ remember: $\left.\iint_{\mathrm{R}^{2}} g(x, y) e^{-i 2 \pi(u x+v)} d x d y\right]$
- Leads to
- the delta function
- model on right

$$
\begin{aligned}
\operatorname{Sample}_{2 \mathrm{D}}(f(x, y)) & =\sum_{i=-\infty}^{\infty} \sum_{j=-\infty}^{\infty} f(x, y) \delta(x-i, y-j) \\
& =f(x, y) \sum_{i=-\infty}^{\infty} \sum_{j=-\infty}^{\infty} \delta(x-i, y-j)
\end{aligned}
$$

## Delta function

- limit to infinity of constant area function:



## A continuous model for a sampled

## function

- We want to be able to
approximate integrals
sensibly
- Leads to
- the delta function
- model on right

$$
\begin{aligned}
\text { Sample }_{2 \mathrm{D}}(f(x, y)) & =\sum_{i=-\infty}^{\infty} \sum_{j=-\infty}^{\infty} f(x, y) \delta(x-i, y-j) \\
& =f(x, y) \sum_{i=-\infty}^{\infty} \sum_{j=-\infty}^{\infty} \delta(x-i, y-j)
\end{aligned}
$$

## The Fourier transform of a sampled signal

$$
\begin{aligned}
F\left(\operatorname{Sample}_{2 \mathrm{D}}(f(x, y))\right) & =F\left(f(x, y) \sum_{i=-\infty}^{\infty} \sum_{i=-\infty}^{\infty} \delta(x-i, y-j)\right) \\
& =F(f(x, y))^{* *} F\left(\sum_{i=-\infty}^{\infty} \sum_{i=-\infty}^{\infty} \delta(x-i, y-j)\right) \\
& =\sum_{i=-\infty}^{\infty} \sum_{j=-\infty}^{\infty} F(u-i, v-j)
\end{aligned}
$$




## Proper sampling

Original
Original
signal


- Low-pass filtering

Low-pass filtered
signal


- Sampling

Sampled
signal

v Sampling

| Reconstruction


## Smoothing as low-pass filtering

- The message of the FT is that high frequencies lead to trouble with sampling.
- Solution: suppress high frequencies before sampling
- multiply the FT of the signal with something that suppresses high frequencies
- or convolve with a low-pass filter
- A filter whose FT is a box is bad, because the filter kernel has infinite support
- Common solution: use a Gaussian
- multiplying FT by Gaussian is equivalent to convolving image with Gaussian.

Sampling without smoothing.
Top row shows the images, sampled at every second pixel to get the next; bottom row shows the magnitude spectrum of these images.


Sampling with smoothing.
Top row shows the images. We get the next image by smoothing the image with a Gaussian with sigma 1 pixel, then sampling at every second pixel to get the next; bottom row shows the magnitude spectrum of these images.


Sampling with smoothing.
Top row shows the images. We get the next image by smoothing the image with a Gaussian with sigma 1.4 pixels, then sampling at every second pixel to get the next; bottom row shows the magnitude spectrum of these images.
$256 \times 256 \quad 128 \times 128 \quad 64 \times 64 \quad 32 \times 32 \quad 16 \times 16$


## Nyquist sampling theorem

- Nyquist theorem: The sampling frequency must be at least twice the highest frequency

$$
\omega_{s} \geq 2 \omega
$$

- If this is not the case the signal needs to be bandlimited before sampling, e.g. with a low-
 pass filter


## ЕНН

## Computation of 2D Fourier Transform

- 2D Fourier Transform can be compute as sequence of 1D Fourier transforms

$$
\begin{aligned}
F(g(x, y))(u, v) & =g(x, y) e^{i 2(u x+v y)} d x d y \\
& =\left(g(x, y) e^{i 2(u x)} d x\right) e^{i 2(v y)} d y
\end{aligned}
$$

- Fast Fourier Transform (FFT) can compute Discrete Fourier Transform very fast (use symmetries) $F=\mathbf{U} f$



## What went wrong?



## ETH

color sampled at half-resolution!

## Signal reconstruction

## Original

Orignal
signal


Low-pass
filtered
filtered
signal




Sampling

Sampled
signal

$\downarrow$ Reconstruction



## Image reconstruction: pixelization

- Who is this?

Harmon \& Julesz 1973


Dali 1976

## Reconstruction filters

Square pixels


Bilinear interpolation


Gaussian reconstruction filter


Perfect reconstruction filter


## Convolution of Box with Box


https://commons.wikimedia.org/wiki/File:Convolution_of_box_signal_with_itself2.gif

## Designing the 'perfect' low-pass filter





## Filtering in Fourier domain



## ETH

## Defocus blurring



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## Motion blurring

Each light dot is transformed into a short line along the $x_{1}$-axis:

$$
h\left(x_{1}, x_{2}\right)=\frac{1}{2 l}\left[\theta\left(x_{1}+l\right)-\theta\left(x_{1}-l\right)\right] \delta\left(x_{2}\right)
$$



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## Image restoration problem

$$
f(\mathbf{x}) \longrightarrow h(\mathbf{x}) \longrightarrow g(\mathbf{x}) \longrightarrow \tilde{h}(\mathbf{x}) \longrightarrow f(\mathbf{x})
$$

The 'inverse' kernel $\tilde{h}(\mathbf{x})$ should compensate the effect of the image degradation $h(\mathbf{x})$, i.e.,

$$
(\tilde{h} * h)(\mathbf{x})=\delta(\mathbf{x})
$$

$\tilde{h}$ may be determined more easily in Fourier space:

$$
\mathcal{F}[\tilde{h}](u, v) \cdot \mathcal{F}[h](u, v)=1
$$

To determine $\mathcal{F}[\tilde{h}]$ we need to estimate

1. the distortion model $h(\mathrm{x})$ (point spread function) or $\mathcal{F}[h](u, v)$ (modulation transfer function)
2. the parameters of $h(\mathbf{x})$, e.g. $r$ for defocussing.

## Image Restoration: Motion Blur

Kernel for motion blur $h(\mathbf{x})=\frac{1}{2 l}\left(\theta\left(x_{1}+l\right)-\theta\left(x_{1}-l\right)\right) \delta\left(x_{2}\right)$
(a light dot is transformed into a small line in $x_{1}$ direction).
Fourier transformation:

$$
\begin{aligned}
\mathcal{F}[h](u, v) & =\frac{1}{2 l} \int_{-l}^{+l} \exp \left(-\imath 2 \pi u x_{1}\right) \underbrace{\int_{-\infty}^{+\infty} \delta\left(x_{2}\right) \exp \left(-\imath 2 \pi v x_{2}\right) d x_{2}}_{=1} d x_{1} \\
& =\frac{\sin (2 \pi u l)}{2 \pi u l}=: \operatorname{sinc}(2 \pi u l)
\end{aligned}
$$

## ETH



## Problems:

- Convolution with the kernel $h$ completely cancels the frequencies $\frac{\nu}{2 l}$ for $\nu \in \mathcal{Z}$. Vanishing frequencies cannot be recovered!
- Noise amplification for $\mathcal{F}[h](u, v) \ll 1$.


## Avoiding noise amplification

## Regularized

reconstruction filter:

$$
\tilde{\mathcal{F}}[\tilde{h}](u, v)=\frac{\mathcal{F}[h]}{|\mathcal{F}[h]|^{2}+\epsilon}
$$

Singularities are avoided by the regularization $\epsilon$.


The size of $\epsilon$ implicitly determines an estimate of the noise level in the image, since we discard signals which are dampened below the size $\epsilon$.

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## Coded Exposure Photography:

 Assisting Motion Deblurring using Fluttered Shutter Raskar, Agrawal, Tumblin (Siggraph2006)Traditional


Image is dark and noisy

Result has Banding Artifacts and some spatial frequencies are lost

Coded


Decoded image is as good as image of a static scene

## Space-time super-resolution

Shechtman et al. PAMI05


## ETH

## Space-time super-resolution

Shechtman et al. PAMI05


## Space-time super-resolution

Shechtman et al. PAMI05

time super-resolution works better than space

(a)

(c)

(b)

Fourier transform of Spatial Blur

(d)

## ЕН

## Spatial super-resolution

- lens+pixel=low-pass filter (desired to avoid aliasing)

- Low-res images $=\mathrm{D}^{*} \mathrm{H}^{*} \mathrm{G}^{*}$ (desired high-res image)
- D: decimate, H:lens+pixel, G: Geometric warp
- Simplified case for translation: $L R=\left(D^{*} G\right)^{*}\left(H^{*} H R\right)$
- G is shift-invariant and commutes with H
- First compute $H^{*} H R$, then deconvolve HR with $H$
- Super-resolution needs to restore attenuated freque
- Many images improve $S / N$ ratio ( $\sim s q r t(n)$ ), which helps


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Eventually Gaussian's double exponential always dominates

# More transforms... 

Eigenfaces


Wavelets


