

4 Transforms

4.2: 2D Linear Filtering

Transforms

4.2

1. 2D Discrete Fourier Transform (2D-DCT)

- Basic properties
- Basic signal transforms

2. 2D Linear Filtering

- Filter design in the spatial domain
- Filter design in the frequency domain

3. Other transforms

- Discrete Cosine Transform (DCT)
- Karhunen-Loeve Transform (KLT)

4. Short-Term Fourier Transform (STFT)

- STFT as a filter bank
- Spectrogram: Time-frequency analysis

Unit Structure

4.2

1. Introduction:

- Image convolution
- Image padding

2. Low-pass filters designed in the spatial domain:

- Example of application: Noise removal
- Trade-off noise removal vs blurring

3. High-pass filters designed in the spatial domain:

- Example of application: Contour detection

4. Filters designed in the frequency domain:

- Ringing effect

5. Summary and Conclusions

Introduction

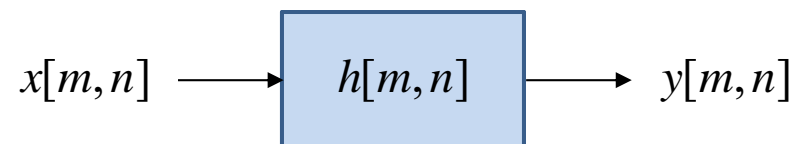
4.2

- Images can be modeled as **linear combinations** of simpler functions; typically:
 - **Impulse functions**, as in the canonical representation.
 - **Complex exponentials**, as in the Fourier representation.
 - ...
- The Fourier representation (and its underlying space/frequency model) is a useful tool to define a natural set of operations such as **Linear Space-Invariant (LSI)** operators.
 - LSI operators linearly combine the pixel values in a given neighborhood of the pixel being processed (**impulse response** or **convolution mask**).
- Linear Space-Invariant (**LSI**) operators can be defined:
 - In the **original (space) domain**, through a **convolution**.
 - In the transformed (frequency) domain, through a **product**.

Image convolution

4.2

A Linear Space-Invariant operator is characterized by its impulse response $h[m, n]$:



and the output can be computed by **convolution**:

$$y[m, n] = \sum_{m'} \sum_{n'} x[m', n'] h[m - m', n - n'] = x[m, n] * h[m, n]$$

$$y[m, n] = \sum_{m'} \sum_{n'} x[m - m', n - n'] h[m', n'] = h[m, n] * x[m, n]$$

In the frequency domain, the output transform is defined by the **product** of the input transform and the **frequency response** $H[k, l]$ of the system.

$$\tilde{y}[m, n] \leftrightarrow Y[k, l] = X[k, l] \cdot H[k, l]$$

if $y[m, n] = x[m, n] * h[m, n]$

Image convolution

4.2

For each pixel position (m, n) :

- A **2D-mirrored version** of $h[m', n']$ ($h[-m', -n']$) is shifted at the pixel position (m, n) ($h[m - m', n - n']$)
- The output $y[m, n]$ is the sum of the pixel values included in the neighborhood defined by the filter and weighted by the co-located filter values ($h[m, n]$ is the **filter impulse response**)

$$y[m, n] = \sum_{m'} \sum_{n'} x[m', n'] h[m - m', n - n']$$

- A **similar interpretation** where the input signal is shifted is also possible

$$y[m, n] = \sum_{m'} \sum_{n'} x[m - m', n - n'] h[m', n']$$

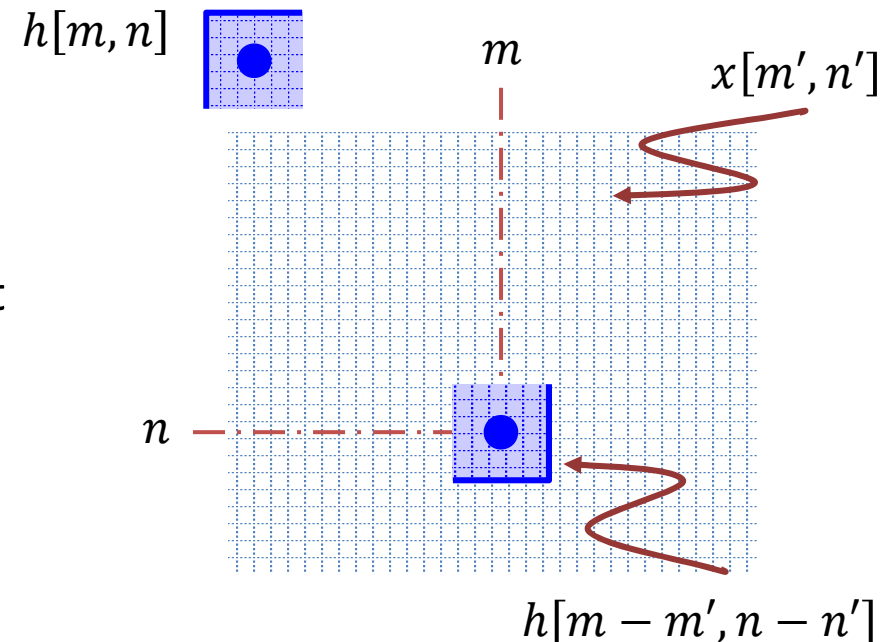


Image padding

4.2

The **computation of the output values close to the image border** requires the use of pixel values outside the input image support.

These values are commonly introduced by:

- **Zero padding:** Including a frame of zeros in the input signal:
 - It may introduce discontinuities in the signal

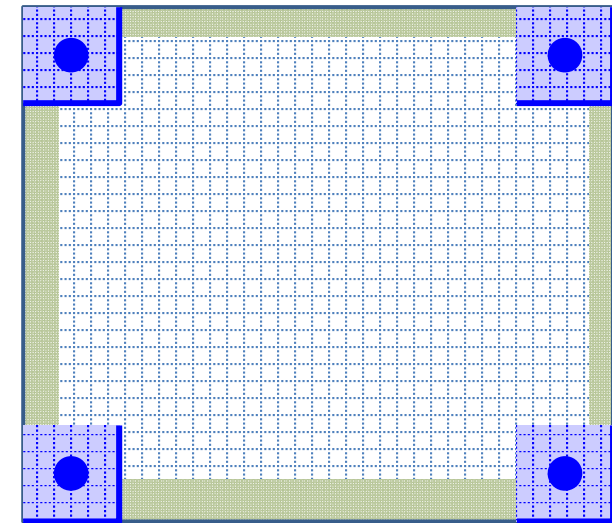
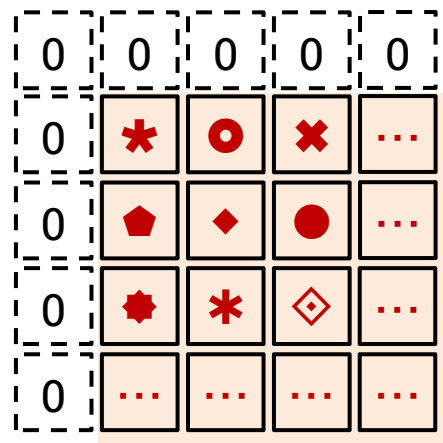


Image padding

4.2

The **computation of the output values close to the image border** requires the use of pixel values outside the input image support.

These values are commonly introduced by:

- **Mirroring:** Including pixel values that are a symmetric mirroring of the input:
 - It generates a smooth transition

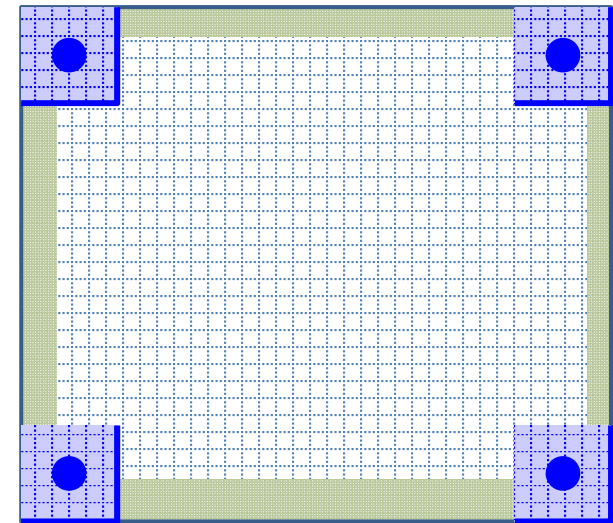
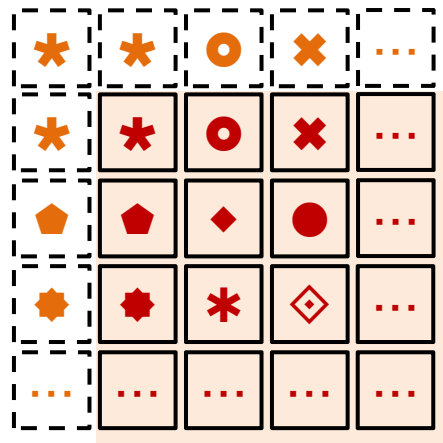


Image padding

4.2

The **computation of the output values close to the image border** requires the use of pixel values outside the input image support.

These values are commonly introduced by:

- **Translation variant filter:** To only use known pixel values:
 - Example with a 3x3 filter

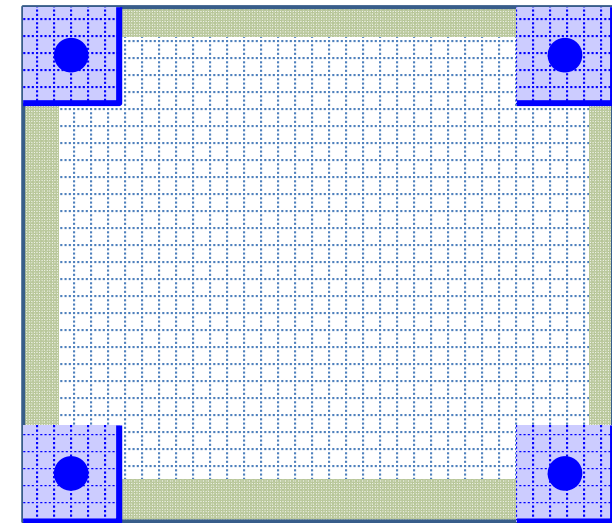
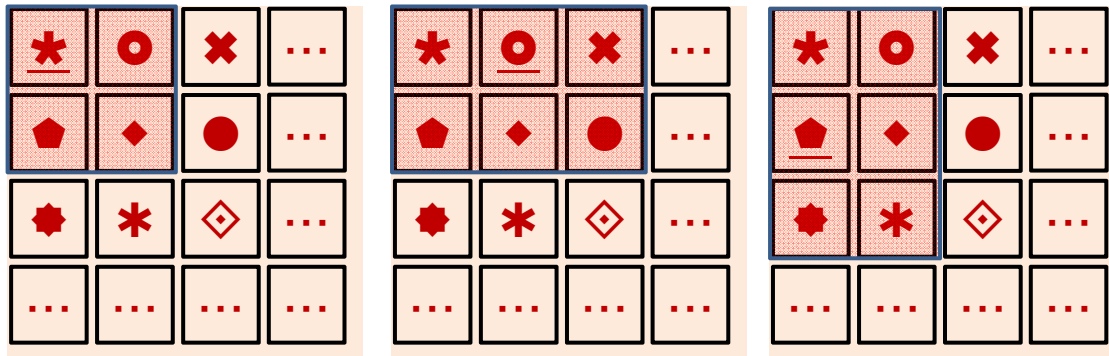
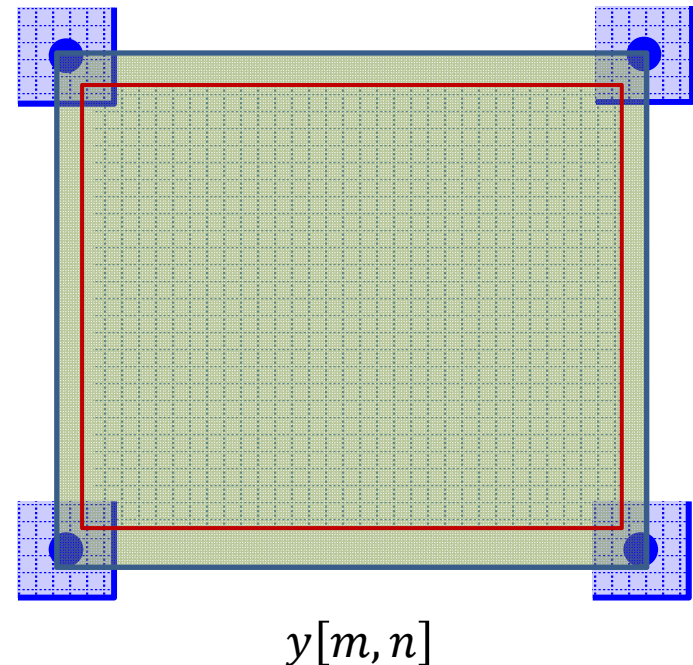


Image convolution and padding

4.2

- The result of the convolution of an image of size $M \times N$ and a filter with impulse response of size $H_1 \times H_2$ is an image of size $(M + H_1 - 1) \times (N + H_2 - 1)$
- Nevertheless, in almost all practical cases, the output image support is **finally restricted** to the original image size.



Separability

4.2

- In some cases, the 2D impulse response can be defined as the product of two 1D impulse responses (**separability**):

$$h[m, n] = h_1[m]h_2[n]$$

- Separability allows **faster implementations**

$$y[m, n] = \sum_{m'} \sum_{n'} x[m', n'] h[m - m', n - n'] = \sum_{m'} \sum_{n'} x[m', n'] h_1[m - m'] h_2[n - n']$$

$$y[m, n] = \sum_{n'} h_2[n - n'] \sum_{m'} x[m', n'] h_1[m - m'] = x[m, n] *_{rows}^{1D} h_1[m] *_{cols}^{1D} h_2[n]$$

- The frequency response of a 2D separable filter can be defined as **the product of the two 1D frequency responses**:

$$H[k, l] = H_1[k]H_2[l]$$

Filters designed in the spatial domain

4.2

- Filters are defined in the spatial domain when their **size** (area of support) is relatively **small**.
 - In other cases, filters are usually defined in the frequency domain.
- Filters are commonly **square, odd in size and symmetric** (3x3, 5x5, ...)
- Two different types of filters are going to be studied:
 - **Low-pass filters**, that perform a spatial average, and their application to noise removal.
 - **High-pass filters**, that perform an estimation of the derivative, and their application to contour detection.

Unit Structure

4.2

1. Introduction:

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- Image padding

2. Low-pass filters designed in the spatial domain:

- Example of application: Noise removal
- Trade-off noise removal vs blurring

3. High-pass filters designed in the spatial domain:

- Example of application: Contour detection

4. Filters designed in the frequency domain:

- Ringing effect

5. Summary and Conclusions

Low pass filters for noise reduction

4.2

We are going to study low-pass filters in the context of **noise reduction**.

In image processing, there are **three main types of noise**:

- **Gaussian noise**: It can appear in the sensors of a CCD camera.

$$p(n) = \frac{1}{\sqrt{2\pi\sigma^2}} e^{-\frac{(n-\mu)^2}{\sigma^2}}$$

- **Uniform noise**: It is produced by the quantization process.

$$p(n) = \begin{cases} \frac{1}{b-a} & \text{if } a \leq n \leq b \\ 0 & \text{otherwise} \end{cases}$$

- **Salt and pepper noise**: Typical of error transmissions or damaged CCD sensors.

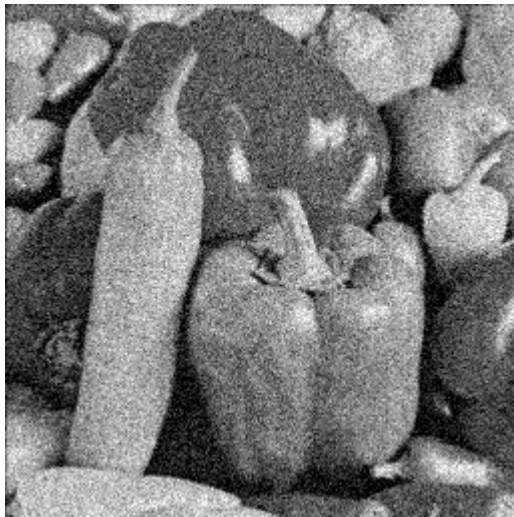
$$\tilde{i}(m,n) = \begin{cases} s & \text{with probability } p \\ i(m,n) & \text{with probability } 1-p \end{cases}$$

s is a random variable that takes values in {black (0), white (255)}

Noise examples

4.2

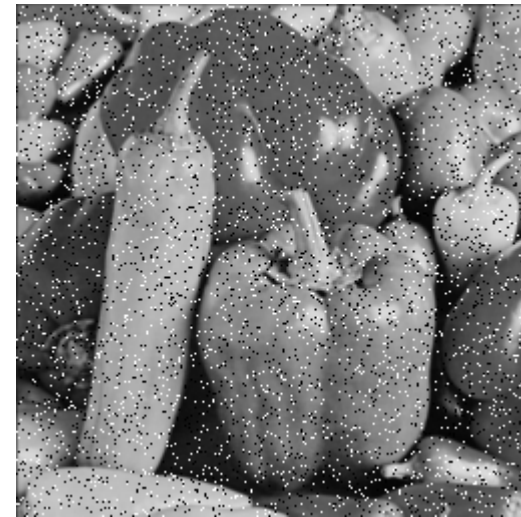
Original image ►



Gaussian noise: ($\mu=0$, $\sigma^2=1$)



Uniform noise



Salt and pepper noise ($p=0.08$)

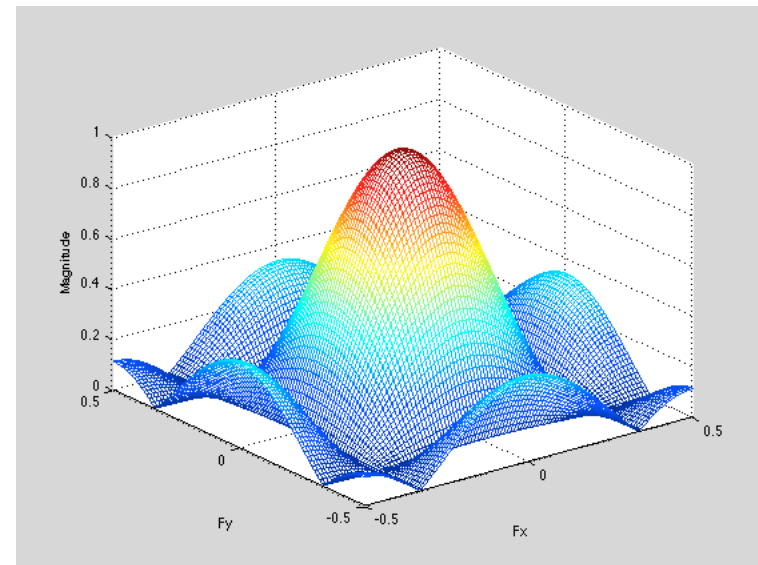
Separability of the average filter

4.2

- The average filter is separable and, therefore, it allows for another type of analysis:

$$h[m, n] = \begin{bmatrix} 1/9 & 1/9 & 1/9 \\ 1/9 & 1/9 & 1/9 \\ 1/9 & 1/9 & 1/9 \end{bmatrix} = h_1[m]h_2[n]$$

$$h_1[m] = \frac{1}{3}\Pi_3[m] \quad h_2[n] = \frac{1}{3}\Pi_3[n]$$



$$H[F_1, F_2] = H_1[F_1]H_2[F_2] \quad \Rightarrow \quad H[F_1, F_2] = \frac{1}{9} \frac{\sin[3\pi F_1]}{\sin[\pi F_1]} \frac{\sin[3\pi F_2]}{\sin[\pi F_2]}$$

Separability of the average filter

4.2

- The average filter is separable and it allows for another type of analysis:

$$H[F_1, F_2] = \frac{1}{9} \frac{\sin[3\pi F_1]}{\sin[\pi F_1]} \frac{\sin[3\pi F_2]}{\sin[\pi F_2]}$$

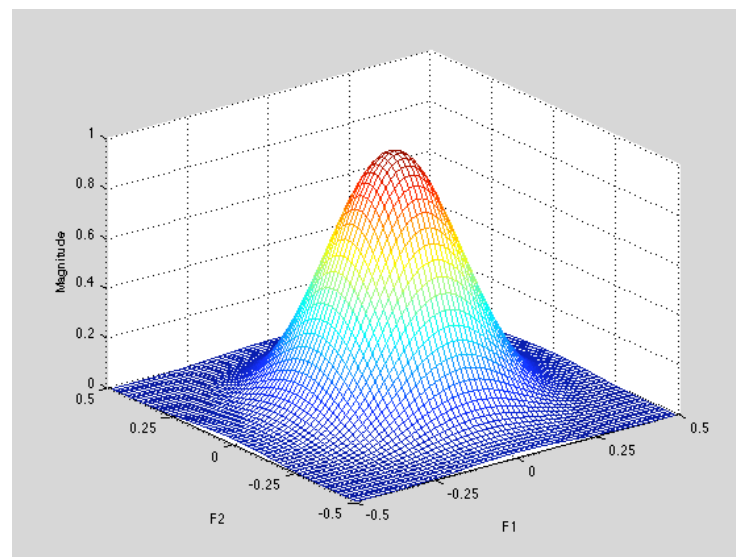
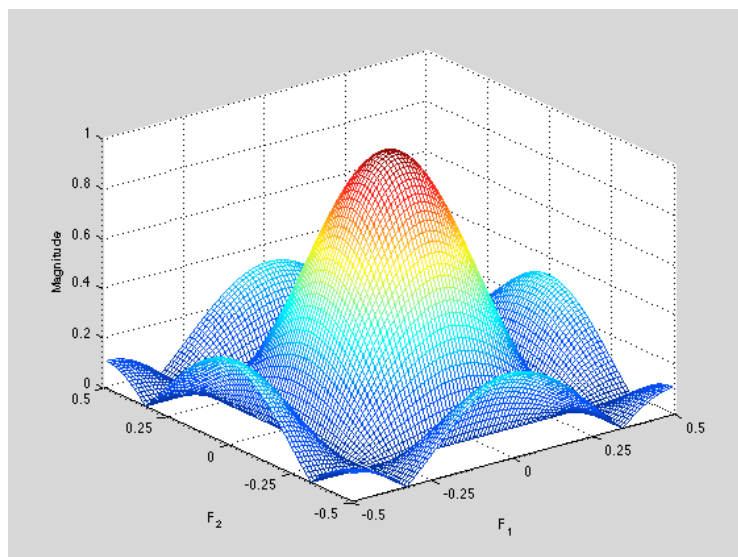
$$\begin{aligned} H_1(F) &= \sum_{m=-\infty}^{\infty} h[m] e^{-i2\pi F m} = \frac{1}{3} \sum_{m=-1}^1 e^{-i2\pi F m} = \\ &= \frac{1}{3} [e^{i2\pi F} + 1 + e^{-i2\pi F}] = \frac{e^{i2\pi F}}{3} [1 + e^{-i2\pi F} + e^{i2\pi 2F}] = \\ &= \left[\sum_{n=0}^{2-1} a^n = \frac{a^0 - a^2}{1 - a} \quad \forall a \neq 1 \right] = \frac{e^{i2\pi F}}{3} \cdot \frac{1 - e^{i2\pi 3F}}{1 - e^{i2\pi F}} = \\ &= \frac{e^{i2\pi F}}{3} \cdot \frac{e^{-i3\pi F}}{e^{i\pi F}} \left[\frac{e^{i8\pi F} - e^{i3\pi F}}{e^{i\pi F} - e^{-i\pi F}} \right] = \frac{1}{3} \frac{\sin 3\pi F}{\sin \pi F} \end{aligned}$$

Spatial averaging filters

4.2

$$\begin{array}{ll} \text{Average 3x3} & h = \begin{bmatrix} 1/9 & 1/9 & 1/9 \\ 1/9 & 1/9 & 1/9 \\ 1/9 & 1/9 & 1/9 \end{bmatrix} \\ \text{Gaussian 5x5} & \sigma=1 \\ h = \begin{bmatrix} 0.00 & 0.01 & 0.02 & 0.01 & 0.00 \\ 0.01 & 0.06 & 0.10 & 0.06 & 0.01 \\ 0.02 & 0.10 & \underline{0.16} & 0.10 & 0.02 \\ 0.01 & 0.06 & 0.10 & 0.06 & 0.01 \\ 0.00 & 0.01 & 0.02 & 0.01 & 0.00 \end{bmatrix} \end{array}$$

Note: In order to preserve the continuous component, the coefficients add up to 1

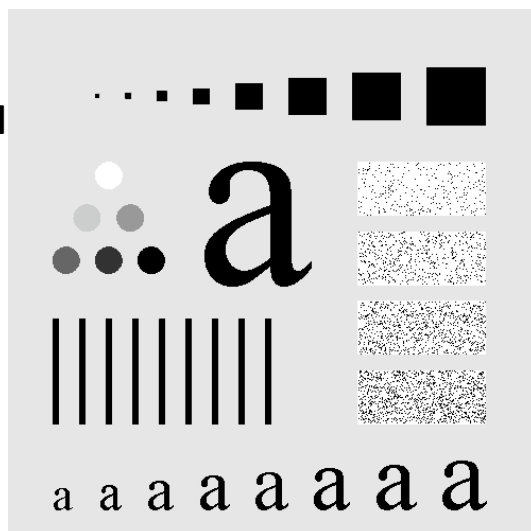


Plots of the frequency response of both filters

Filtering examples

4.2

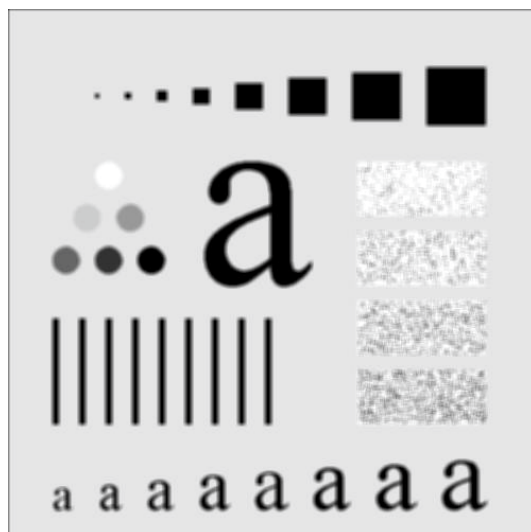
Original
image



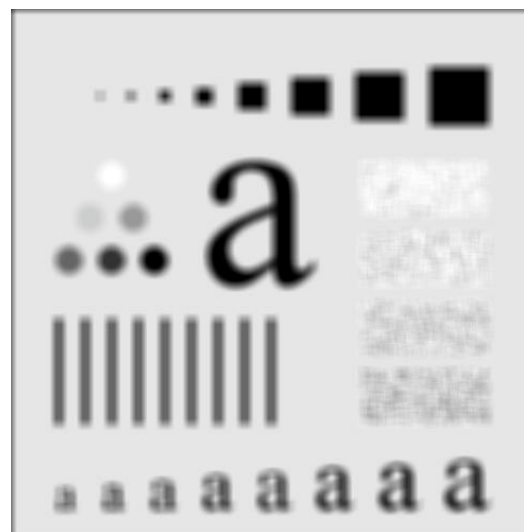
Gaussian
 $\sigma=4$



Average
5x5



Average
11x11



Example of noise reduction

4.2

Original
image



Uniform
noise

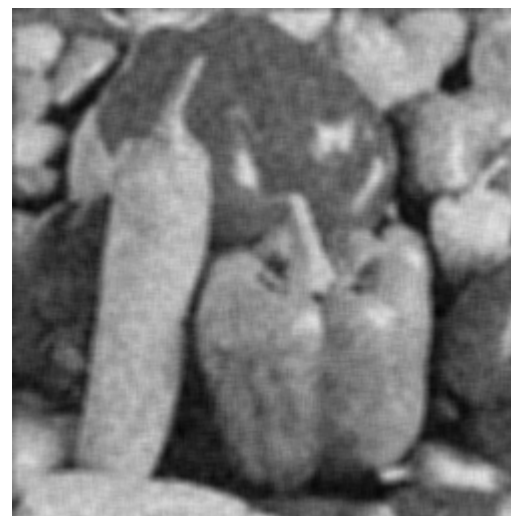


Averaging
filter

$$\begin{bmatrix} 1/9 & 1/9 & 1/9 \\ 1/9 & \underline{1/9} & 1/9 \\ 1/9 & 1/9 & 1/9 \end{bmatrix}$$

Gaussian
Filter ($\sigma=0.5$)

$$\begin{bmatrix} 0.011 & 0.084 & 0.011 \\ 0.084 & \underline{0.619} & 0.084 \\ 0.011 & 0.084 & 0.011 \end{bmatrix}$$



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5. Summary and Conclusions

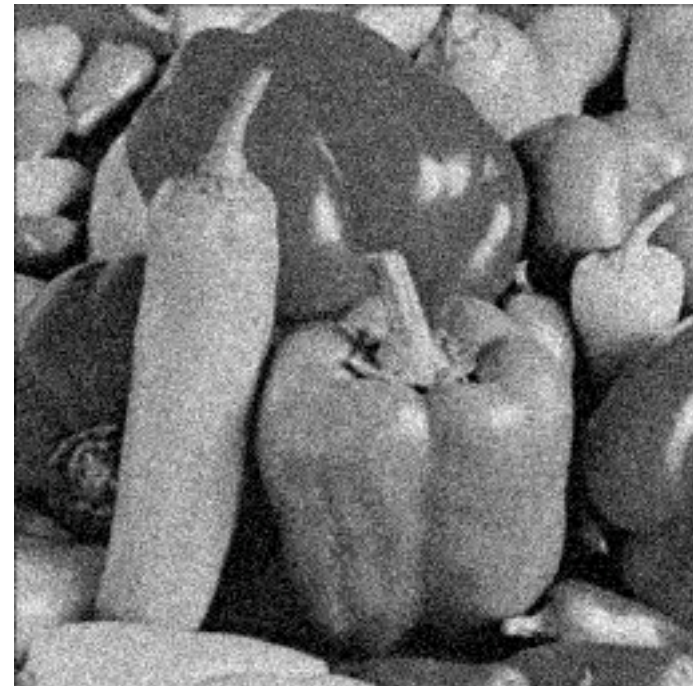
Filter design

4.2

Original image



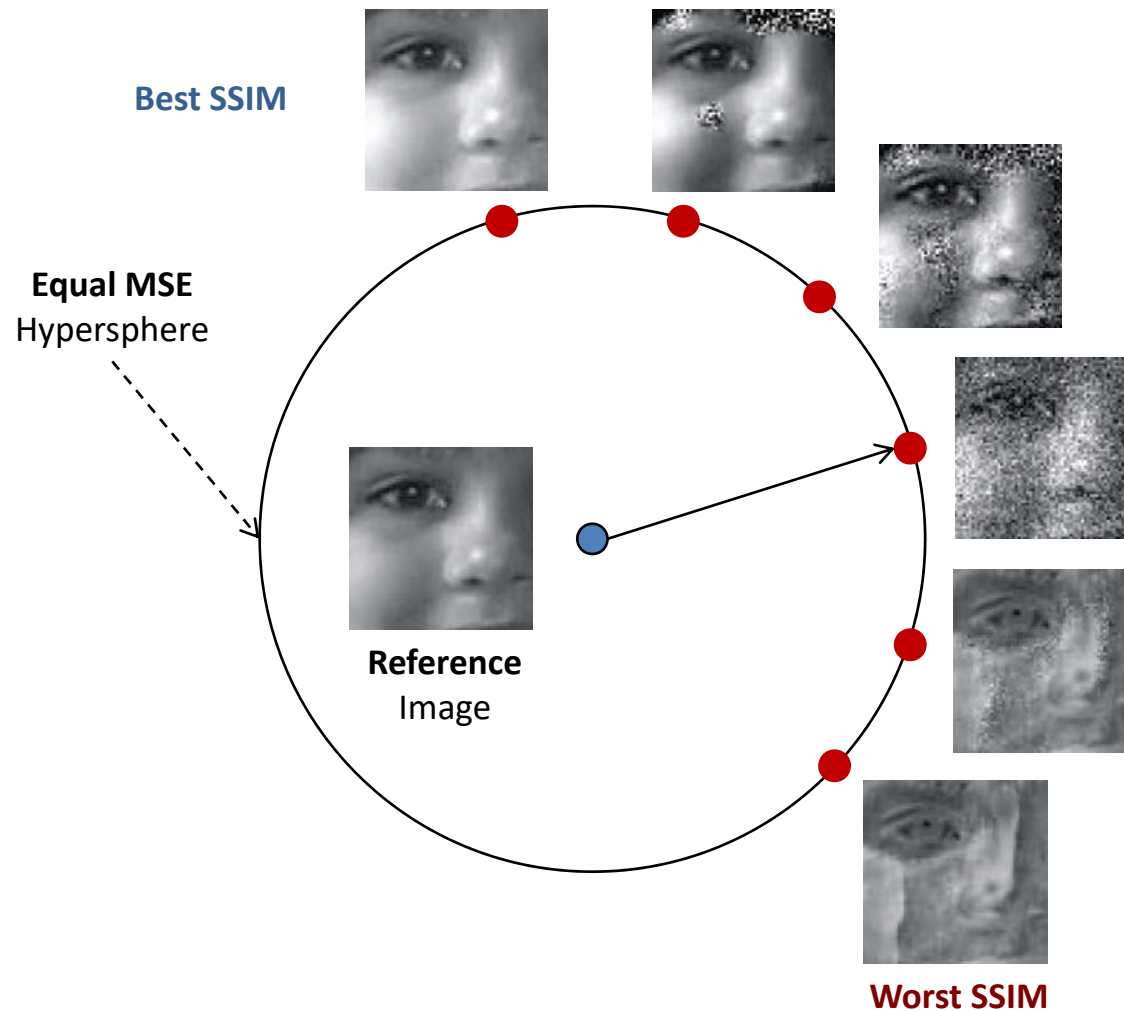
**Image corrupted by
additive Gaussian noise**



**PSNR=27,88 dB
SSIM=0,448**

User oriented assessment

4.2



Objective and perceptual criteria for human:

Researchers are continuously looking for objective measures that model better the subjective behavior of the Human Visual System.

For instance, the **Structural SIMilarity Index (SSIM)**.

Z. Wang and A.C. Bovik, "Mean Square Error: Love it or Leave it?"
IEEE Signal Processing Magazine, pp. 98 - 117, January 2009.

Filter design

4.2

Assume we use an averaging filter: how to define the size of the **impulse response** ?

- Small impulse responses: do not remove much noise
- Large windows: blur edges



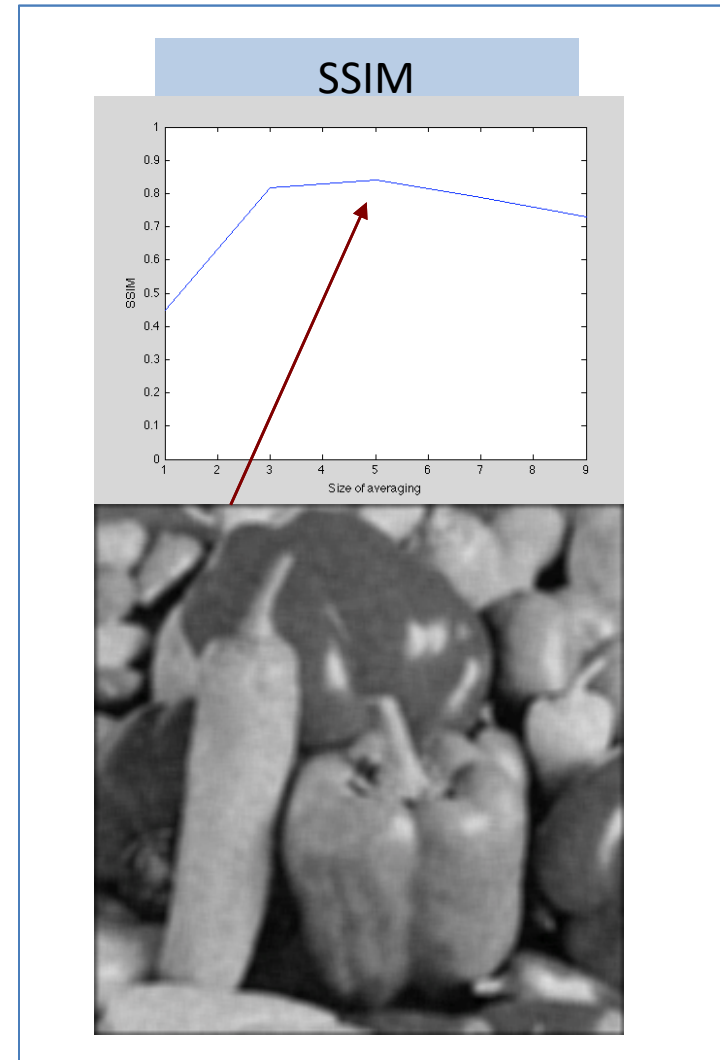
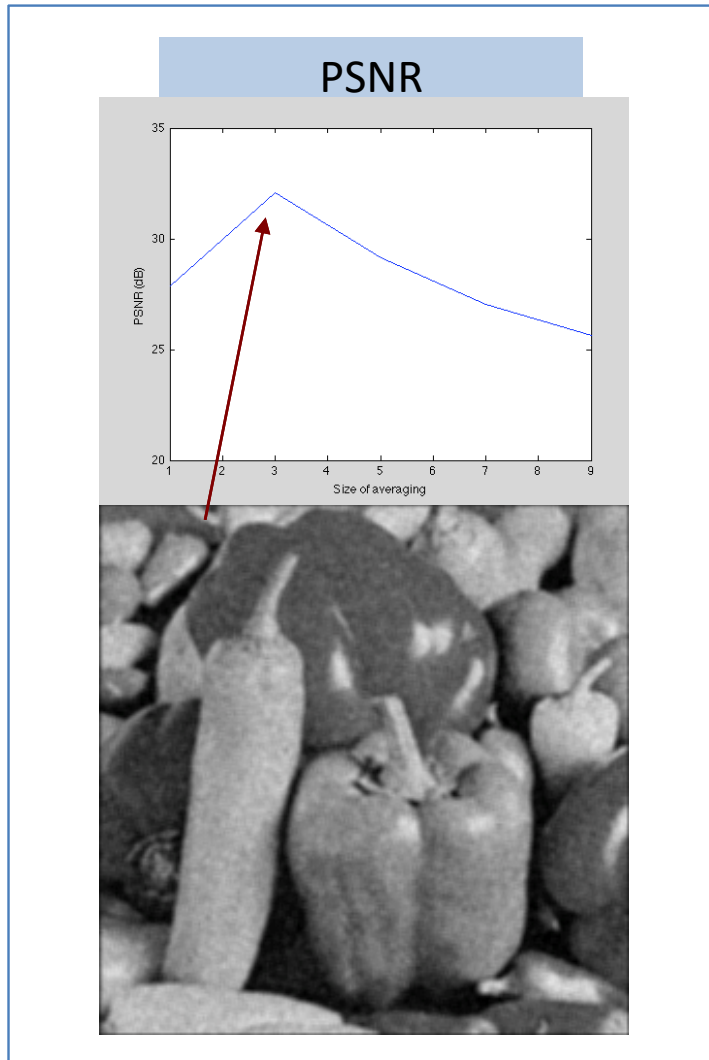
Short window



Large window

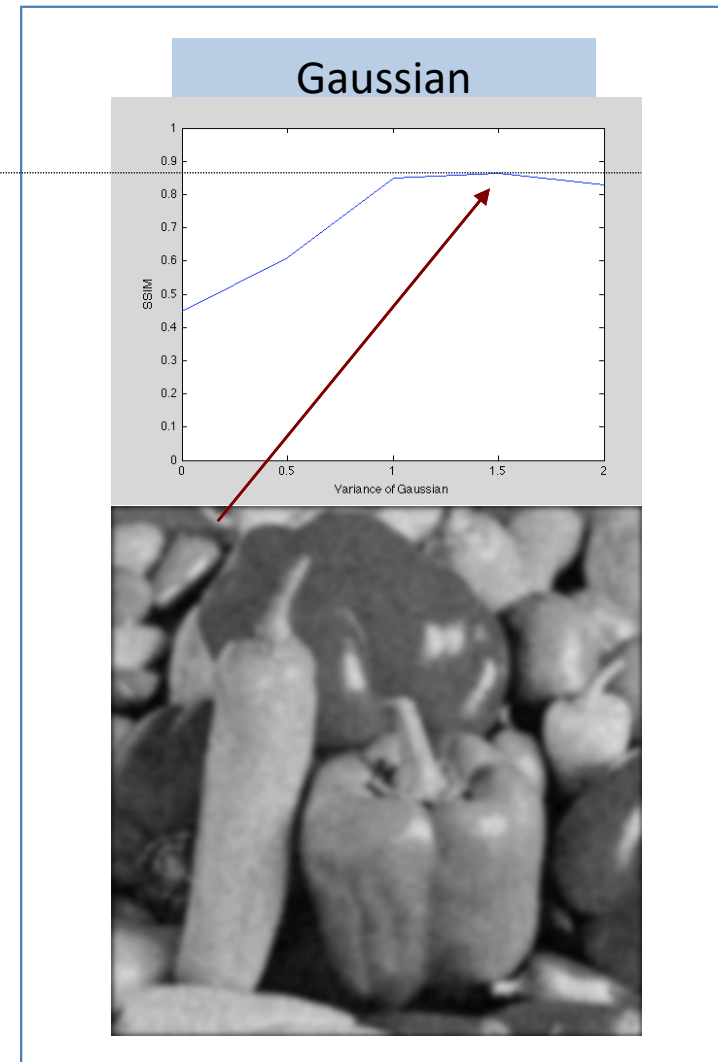
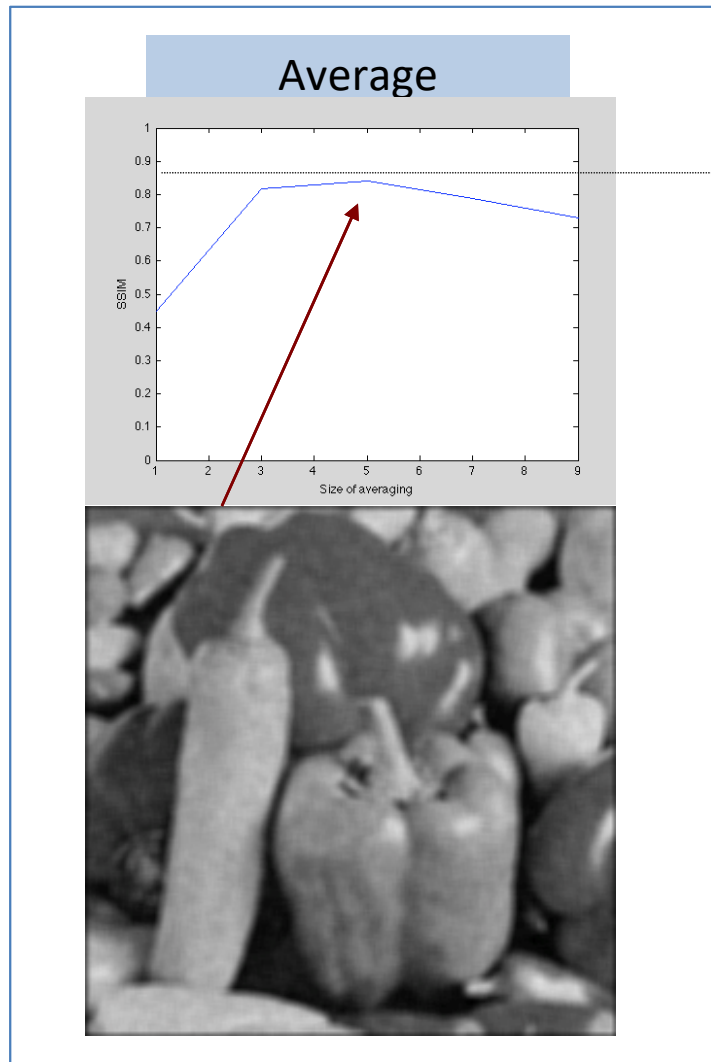
Trade-off: Noise reduction vs Blurring

4.2



Averaging vs Gaussian filter

4.2



Impulsive noise (salt and pepper)

4.2

Original image



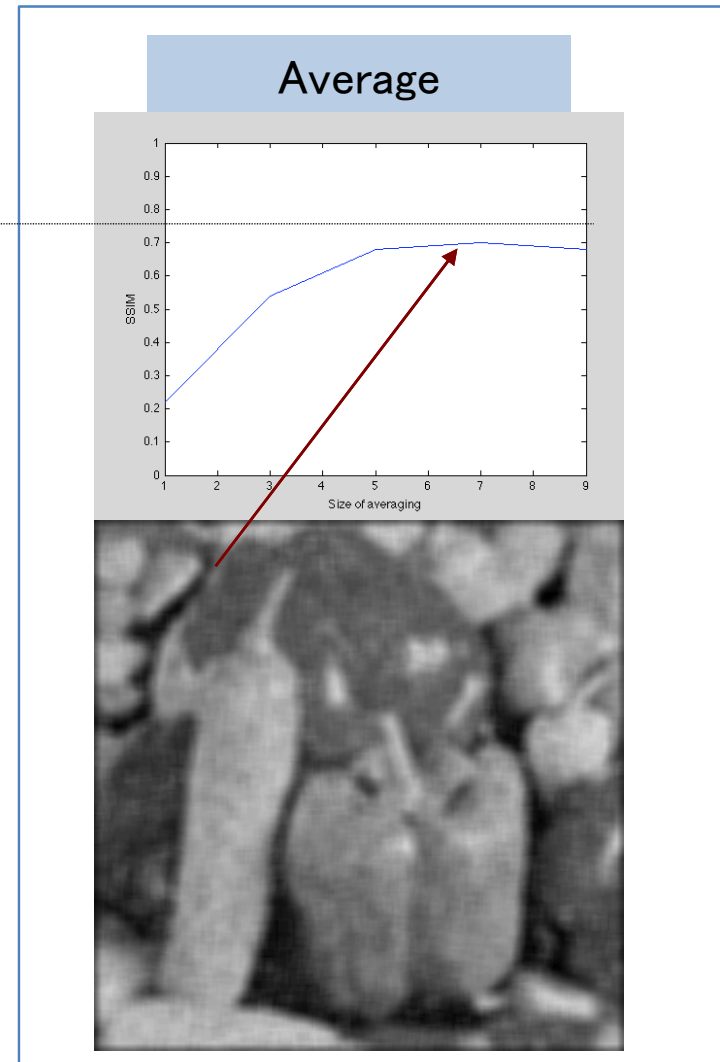
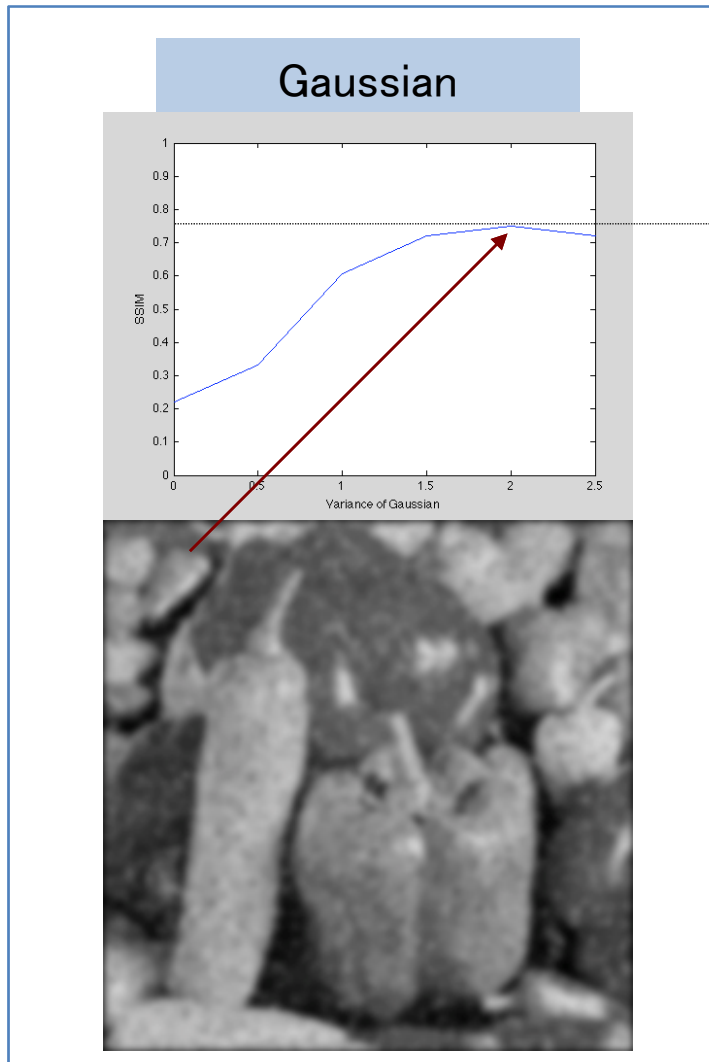
Image corrupted by
impulsive noise



PSNR=19,22 dB
SSIM=0,22

Impulsive (salt and pepper) noise

4.2



Impulsive (salt and pepper) noise

4.2

The Gaussian filter is better than the Averaging filter.... **But can we do better?**

(Original noisy image: PSNR=19,22 dB; SSIM=0,22)



Gaussian

PSNR=27,50 dB SSIM=0,75



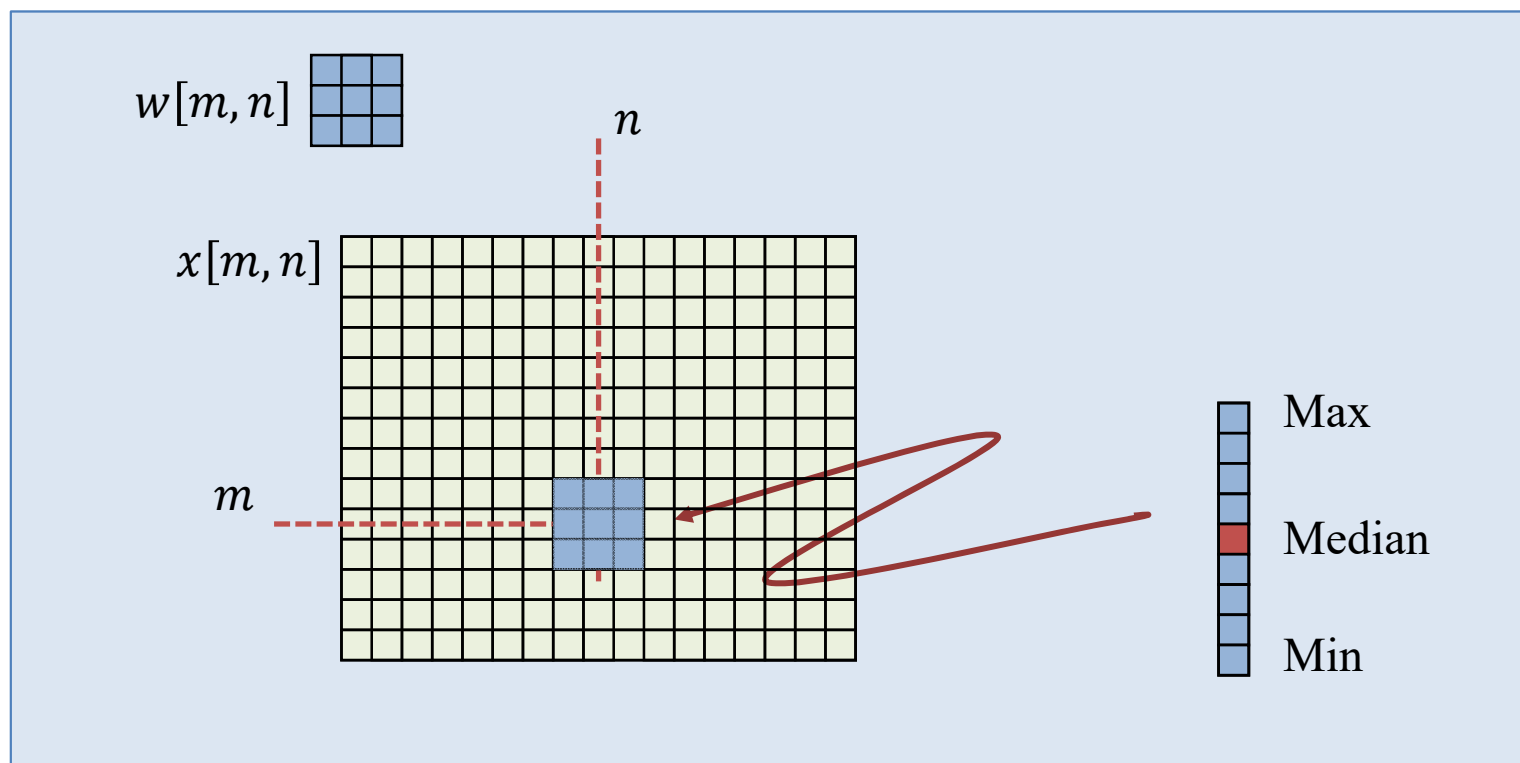
?

PSNR=34,91 dB SSIM=0,97

Impulsive noise (salt and pepper)

4.2

Median filter: a non linear filter that is very efficient for impulsive noise removal!



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4.2

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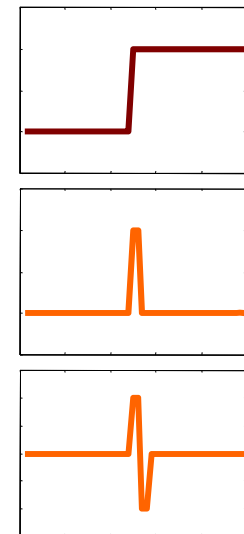
Contour detection

4.2

- We are going to study high-pass filters in the context of **contour detection**.

Contours (**transitions**) in a signal can be detected by means of :

- First derivative:** There is a **local maximum** (or minimum) in the derivative in the position of the transition.
- Second derivative:** There is a **zero crossing** in the second derivative in the position of the transition.



- In 2D, the **gradient** or the **Laplacian** of the image are studied:

$$\nabla f(x, y) = \left[\frac{\partial f(x, y)}{\partial x}, \frac{\partial f(x, y)}{\partial y} \right]^T$$

$$\Delta f(x, y) = \nabla^2 f(x, y) = \frac{\partial^2 f(x, y)}{\partial x^2} + \frac{\partial^2 f(x, y)}{\partial y^2}$$

Gradient approximation

4.2

- In image processing, gradients can be approximated by means of **finite differences equations**, one for each direction (x, y).

$$\frac{\partial f(x, y)}{\partial x} = \lim_{\Delta x \rightarrow 0} \left[\frac{f(x + \Delta x, y) - f(x, y)}{\Delta x} \right] \approx \frac{f(x + \Delta x, y) - f(x, y)}{\Delta x} \quad \text{Forward difference}$$

$$\frac{\partial f(x, y)}{\partial x} = \lim_{\Delta x \rightarrow 0} \left[\frac{f(x, y) - f(x - \Delta x, y)}{\Delta x} \right] \approx \frac{f(x, y) - f(x - \Delta x, y)}{\Delta x} \quad \text{Backward difference}$$

$$\frac{\partial f(x, y)}{\partial x} = \lim_{\Delta x \rightarrow 0} \left[\frac{f(x + \Delta x, y) - f(x - \Delta x, y)}{2\Delta x} \right] \approx \frac{f(x + \Delta x, y) - f(x - \Delta x, y)}{2\Delta x} \quad \text{Symmetric difference}$$

$$\begin{array}{cc} \frac{\partial f(x, y)}{\partial x} \bigg|_{x=m, y=n} \approx f(m+1, n) - f(m, n) & \frac{\partial f(x, y)}{\partial x} \bigg|_{x=m, y=n} \approx f(m, n) - f(m-1, n) \\ \frac{\partial f(x, y)}{\partial x} \bigg|_{x=m, y=n} \approx \frac{f(m+1, n) - f(m-1, n)}{2} \end{array}$$

Gradient approximation

4.2

- Each of these equations can be implemented as an **LTI filter**.

$$\left. \frac{\partial f(x, y)}{\partial x} \right|_{\substack{x=m \\ y=n}} \approx f(m+1, n) - f(m, n)$$

$$h_x^+[m] = [1, \underline{-1}] \Rightarrow \frac{\partial f}{\partial x} \approx f * h_x^+$$

$$\left. \frac{\partial f(x, y)}{\partial x} \right|_{\substack{x=m \\ y=n}} \approx f(m, n) - f(m-1, n)$$

$$h_x^-[m] = [\underline{1}, -1] \Rightarrow \frac{\partial f}{\partial x} \approx f * h_x^-$$

$$\left. \frac{\partial f(x, y)}{\partial x} \right|_{\substack{x=m \\ y=n}} \approx \frac{f(m+1, n) - f(m-1, n)}{2}$$

$$h_x^s[m] = [1, \underline{0}, -1] \Rightarrow \frac{\partial f}{\partial x} \approx f * h_x^s$$

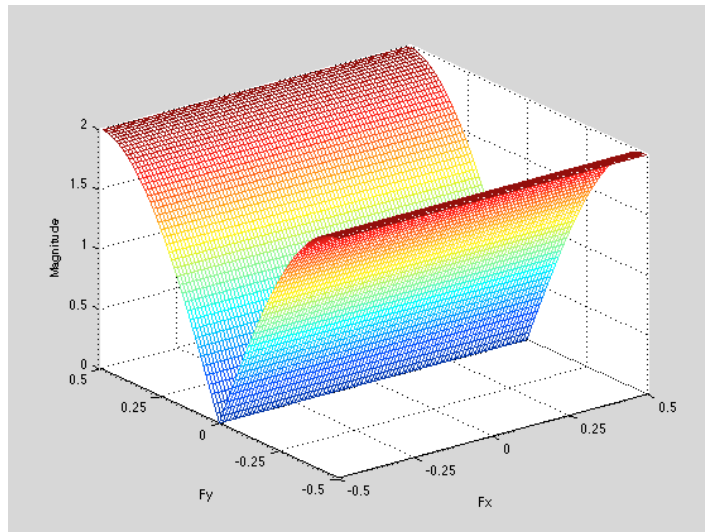
- In the symmetric filter, **the factor ½ has been removed** since usually the exact value of the gradient is not sought.
- Analogous filters can be defined in the **vertical direction**.

Gradient approximation

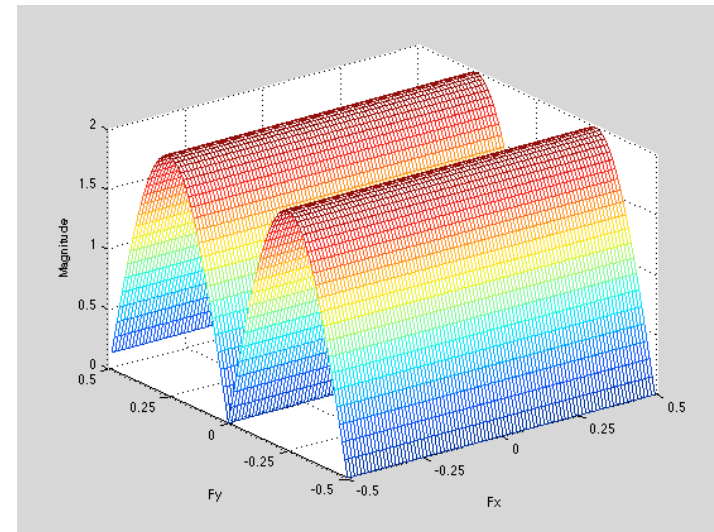
4.2

- Frequency response of the basic derivative approximations

$$h_y^-[m,n] = \begin{bmatrix} 1 \\ -1 \end{bmatrix}$$



$$h_y^s[m,n] = \begin{bmatrix} 1 \\ 0 \\ -1 \end{bmatrix}$$



Gradient approximation

4.2

- Frequency response of the basic derivative approximations

$$h[m, n] = \begin{bmatrix} \frac{1}{2} \\ -\frac{1}{2} \end{bmatrix} \quad \left\{ \begin{array}{l} \Rightarrow \begin{bmatrix} 0 & 0 & 0 \\ 0 & 1 & 0 \\ 0 & -1 & 0 \end{bmatrix} \Rightarrow \begin{array}{l} h_1[m] = \delta[m] \\ h_2[n] = \delta[n] - \delta[n-1] \end{array} \end{array} \right.$$

$$h_2[n] = \delta[n] - \delta[n-1] \Rightarrow H_2(f_2) = 1 - e^{-i2\pi f_2}$$

$$H_2(f_2) = e^{-i\pi f_2} [e^{i\pi f_2} - e^{-i\pi f_2}] = 2i e^{-i\pi f_2} \sin \pi f_2$$

$$|H(f_1, f_2)| = |h_1(f_1) \cdot H_2(f_2)| = 2 |\sin \pi f_2| \quad \text{FIRST CASE}$$

$$|H(f_1, f_2)| = |h_1(f_1) \cdot H_2(f_2)| = 2 |\sin 2\pi f_2| \quad \text{SECOND CASE}$$

Gradient approximation

4.2

- There exist **several implementations** of the filter that approximates the gradient in a given direction. Let us denote them, in general, as h_x (h_y).
- Typically, these filters do not estimate the gradient relying only on the data in one row (column) but in a **set of neighbor rows** (columns)
- This allows averaging among several rows (columns) and being more **robust against the presence of noise**

$$h_x = \begin{bmatrix} 1 & 0 & -1 \\ 1 & \underline{0} & -1 \\ 1 & 0 & -1 \end{bmatrix}$$

Horizontal derivative
Detects vertical contours

$$h_y = \begin{bmatrix} 1 & 1 & 1 \\ 0 & \underline{0} & 0 \\ -1 & -1 & -1 \end{bmatrix}$$

Vertical derivative
Detects horizontal contours

Gradient approximation

4.2

Let us see how the filter works in an image showing a perfectly defined contour.

Note: We use zero padding to study its effect

$$h_x[m,n] = \begin{bmatrix} 1 & 0 & -1 \\ 1 & 0 & -1 \\ 1 & 0 & -1 \end{bmatrix} \Rightarrow h_x[-m,-n] = \begin{bmatrix} -1 & 0 & 1 \\ -1 & 0 & 1 \\ -1 & 0 & 1 \end{bmatrix}$$

Horizontal derivative
Detects vertical contours

0	0	0	0	0	0	0	0	0
0	0	...	0	1	1	...	1	0
0	0	...	0	1	1	...	1	0
0	0
0	0	...	0	1	1	...	1	0
0	0	...	0	1	1	...	1	0
0	0	0	0	0	0	0	0	0

0	0	0	0	0	0	0	0	0
0	0	...	2	2	0	...	-2	0
0	0	...	3	3	0	...	-3	0
...
0	0	...	3	3	0	...	-3	0
0	0	...	2	2	0	...	-2	0
0	0	0	0	0	0	0	0	0

Gradient approximation

4.2

Let us see how the filter works in an image showing a perfectly defined contour.

Note: We use zero padding to study its effect

$$h_y[m,n] = \begin{bmatrix} 1 & 1 & 1 \\ 0 & 0 & 0 \\ -1 & -1 & -1 \end{bmatrix} \Rightarrow h_y[-m,-n] = \begin{bmatrix} -1 & -1 & -1 \\ 0 & 0 & 0 \\ 1 & 1 & 1 \end{bmatrix}$$

Vertical derivative
Detects horizontal contours

0	0	...	0	0	0	...	0	0
0	0	...	0	1	1	...	1	0
0	0	...	0	1	1	...	1	0
...
0	0	...	0	1	1	...	1	0
0	0	...	0	1	1	...	1	0
0	0	...	0	0	0	...	0	0

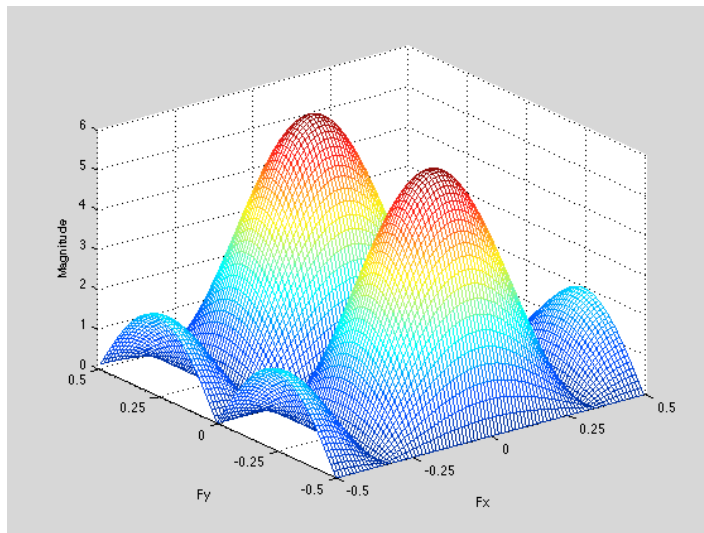
0	0	0	0	0	0	0	0	0
0	0	...	1	2	3	...	2	0
0	0	...	0	0	0	...	0	0
0	0
0	0	...	0	0	0	...	0	0
0	0	...	-1	-2	-3	...	-2	0
0	0	0	0	0	0	0	0	0

Gradient approximation

4.2

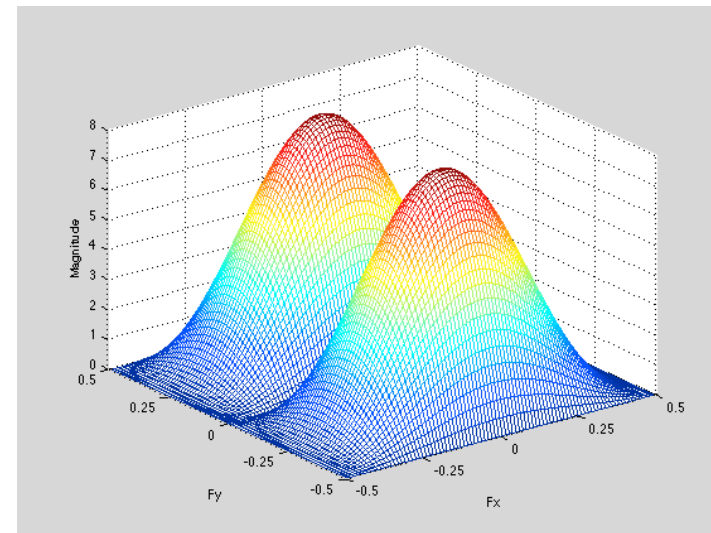
Prewitt

$$h_y[m,n] = \begin{bmatrix} 1 & 1 & 1 \\ 0 & 0 & 0 \\ -1 & -1 & -1 \end{bmatrix}$$



Sobel

$$h_y[m,n] = \begin{bmatrix} 1 & 2 & 1 \\ 0 & 0 & 0 \\ -1 & -2 & -1 \end{bmatrix}$$



- ❑ Compute the frequency responses of these filters

Gradient approximation

4.2

- Compute the frequency responses of these filters

PREWITT

$$h_y[m, n] = \begin{bmatrix} 1 & 1 & 1 \\ 0 & 0 & 0 \\ -1 & -1 & -1 \end{bmatrix} = \begin{bmatrix} 1 \\ 0 \\ -1 \end{bmatrix} [1 \ 1 \ 1] = h_{y_1}[m] h_{y_2}[n]$$

where $h_{y_1}[n]$: GRADIENT APPROX. $h_{y_2}[m]$: AVERAGE

Sobel

$$h_y[m, n] = \begin{bmatrix} 1 & 2 & 1 \\ 0 & 0 & 0 \\ -1 & -2 & -1 \end{bmatrix} = \begin{bmatrix} 1 \\ 0 \\ -1 \end{bmatrix} [1 \ 2 \ 1] = h_{y_1}[m] h_{y_2}[n]$$

where $h_{y_1}[n]$: GRADIENT APPROX. $h_{y_2}[m]$: TRIANGLE

Gradient approximation

4.2

Given the previous filters, a set of images can be defined that help analyzing the gradient of the input image:

$$\nabla f[m, n] = \begin{bmatrix} g_x[m, n] \\ g_y[m, n] \end{bmatrix}$$

Estimation of the gradient

$$g_x[m, n] = f[m, n] * h_x[m, n]$$

Image that stores the estimation of the horizontal gradient

$$g_y[m, n] = f[m, n] * h_y[m, n]$$

Image that stores the estimation of the vertical gradient

$$|\nabla f[m, n]| = \sqrt{g_x^2[m, n] + g_y^2[m, n]}$$

Image that stores the estimation of the gradient magnitude

$$\theta_{\nabla f}[m, n] = \arctan \left[\frac{g_y[m, n]}{g_x[m, n]} \right]$$

Image that stores the estimation of the gradient phase

Examples

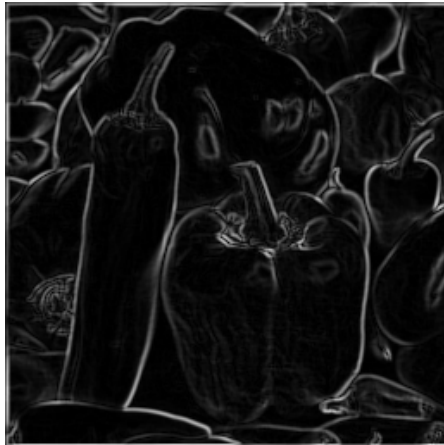
4.2

The **Prewitt filter** detects both horizontal and vertical contours and it is robust with respect to the noise.

Original
image

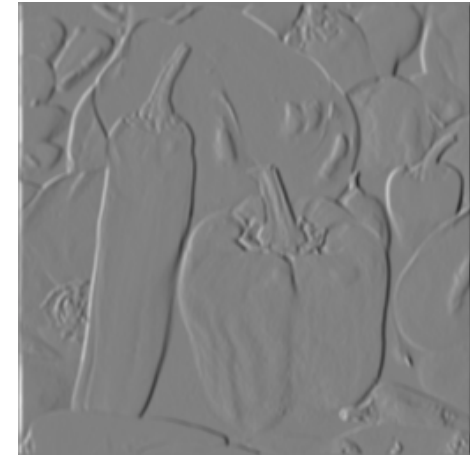


Gradient
magnitude



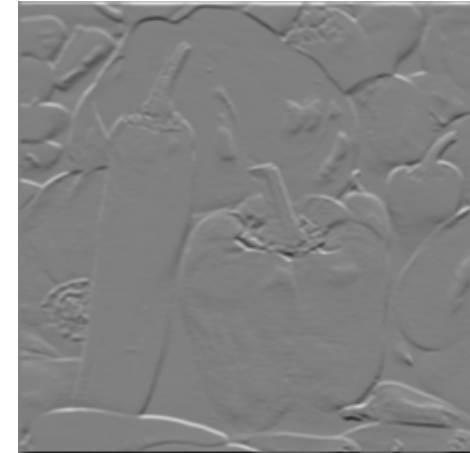
Horizontal derivative
Detects vertical contours

$$h_x = \begin{bmatrix} 1 & 0 & -1 \\ 1 & 0 & -1 \\ 1 & 0 & -1 \end{bmatrix}$$



Vertical derivative
Detects horizontal contours

$$h_y = \begin{bmatrix} 1 & 1 & 1 \\ 0 & 0 & 0 \\ -1 & -1 & -1 \end{bmatrix}$$



Examples

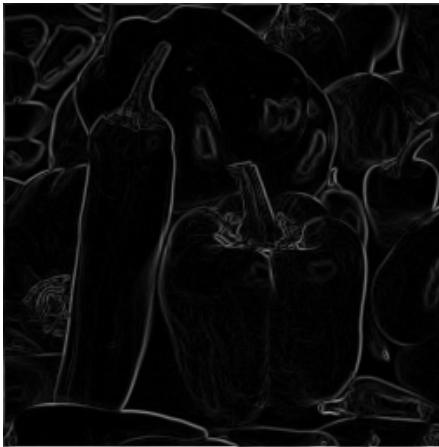
4.2

The **Roberts filter** detects contours in both diagonal directions but it is not robust with respect to the noise.

Original
image

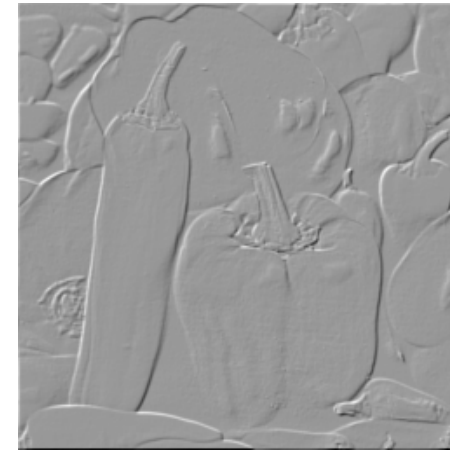


Gradient
magnitude



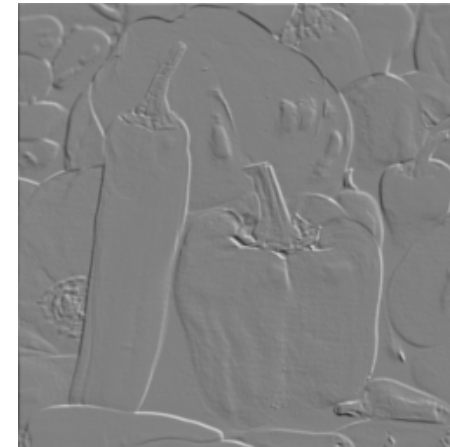
Diagonal direction
Detects contours at 45°

$$h_1 = \begin{bmatrix} 1 & 0 \\ 0 & -1 \end{bmatrix}$$



Diagonal direction
Detects contours at 135°

$$h_2 = \begin{bmatrix} 0 & 1 \\ -1 & 0 \end{bmatrix}$$



Examples

4.2

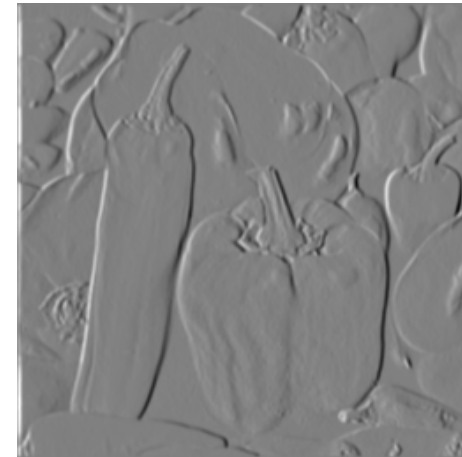
The **Sobel filter** detects both horizontal and vertical contours and increases the robustness with respect to the noise.

Original
image

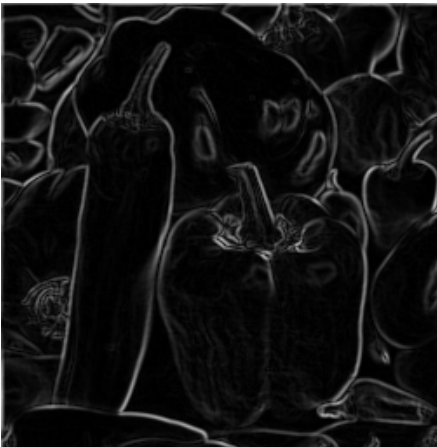


Horizontal derivative
Detects vertical contours

$$h_x[m,n] = \begin{bmatrix} -1 & 0 & 1 \\ -2 & 0 & 2 \\ -1 & 0 & 1 \end{bmatrix}$$

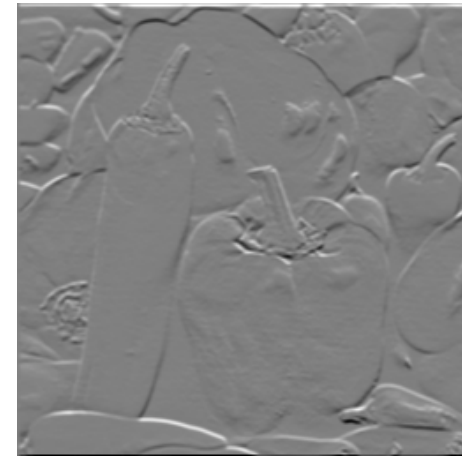


Gradient
magnitude



Vertical derivative
Detects horizontal contours

$$h_y[m,n] = \begin{bmatrix} 1 & 2 & 1 \\ 0 & 0 & 0 \\ -1 & -2 & -1 \end{bmatrix}$$



Laplacian approximation

4.2

- In image processing, the Laplacian operator can be approximated by means of **equations in finite differences**.

$$\begin{aligned} \left. \frac{\partial^2 f(x, y)}{\partial x^2} \right|_{x=m, y=n} &\approx f(m+1, n) - 2f(m, n) + f(m-1, n) \\ \left. \frac{\partial^2 f(x, y)}{\partial y^2} \right|_{x=m, y=n} &\approx f(m, n+1) - 2f(m, n) + f(m, n-1) \end{aligned} \quad \xrightarrow{\quad} \quad \Delta f = \frac{\partial^2 f}{\partial x^2} + \frac{\partial^2 f}{\partial y^2}$$

$$\Delta f(x, y) \Big|_{x=m, y=n} = \frac{\partial^2 f}{\partial x^2} + \frac{\partial^2 f}{\partial y^2} \Big|_{x=m, y=n} \approx f(m+1, n) + f(m-1, n) + f(m, n+1) + f(m, n-1) - 4f(m, n)$$

- This equation can be implemented by means of the convolution with the so-called **Laplacian filter**:

$$h = \begin{bmatrix} 0 & 1 & 0 \\ 1 & -4 & 1 \\ 0 & 1 & 0 \end{bmatrix} \quad \text{or} \quad h = \begin{bmatrix} 0 & -1 & 0 \\ -1 & 4 & -1 \\ 0 & -1 & 0 \end{bmatrix}$$

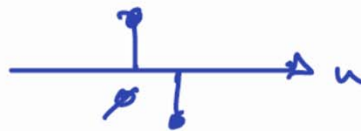
Laplacian approximation

4.2

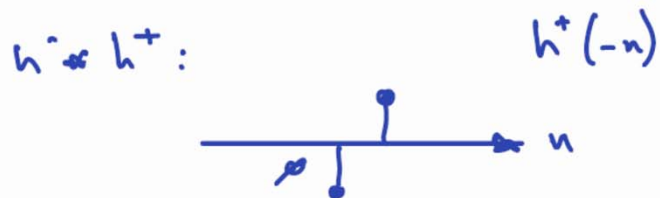
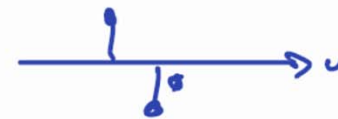
- The Laplacian operator is approximated by equations in finite differences.

$$\left. \frac{\partial^2 f(x, y)}{\partial x^2} \right|_{\substack{x=m \\ y=n}} \approx f(m+1, n) - 2f(m, n) + f(m-1, n)$$

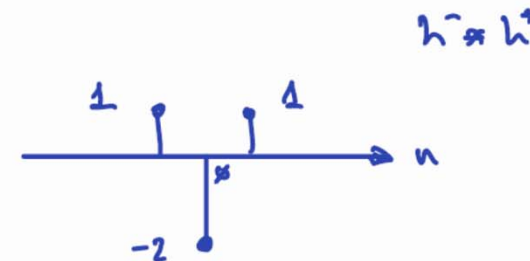
$$h^- : \delta[n] - \delta[n-1]$$



$$h^+ : \delta[n+1] - \delta[n]$$



\Rightarrow

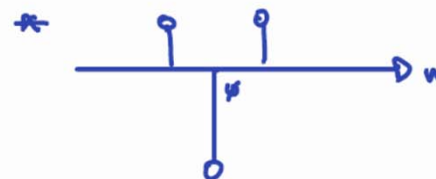


THE FILTERING IS :

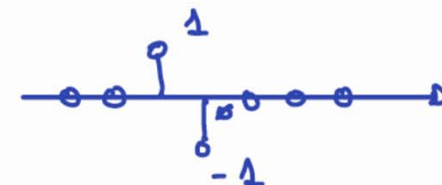
$x[n]$



$h[n]$



$y[n]$



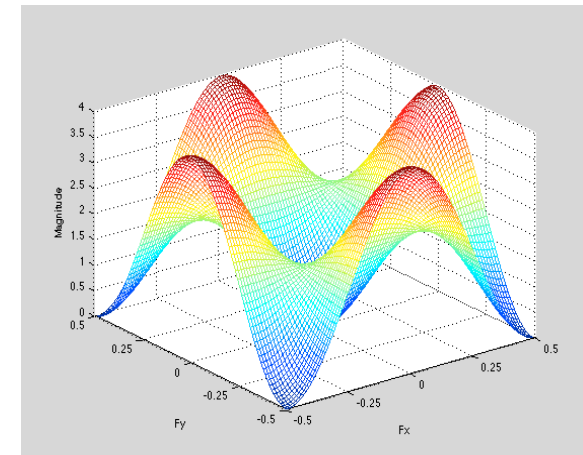
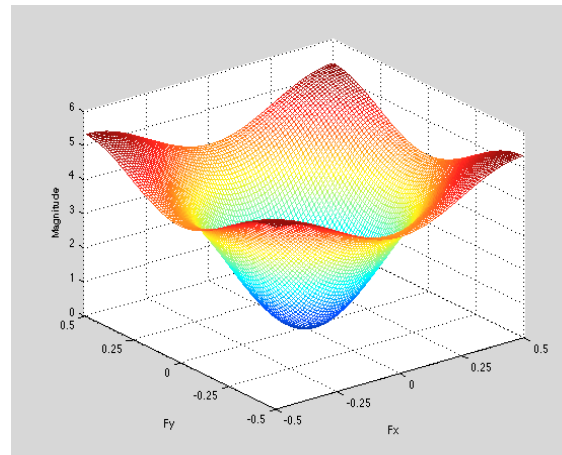
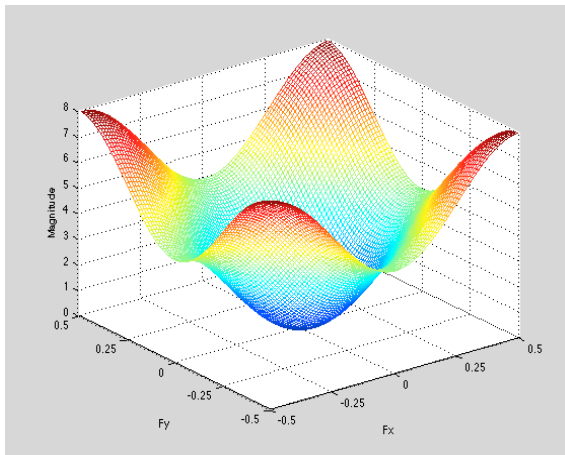
Laplacian approximation

4.2

Several approximations can be used to get a more or less isotropic response.

$$h = \begin{bmatrix} 0 & 1 & 0 \\ 1 & -4 & 1 \\ 0 & 1 & 0 \end{bmatrix} \xrightarrow{2/3} h = \begin{bmatrix} 1/6 & 2/3 & 1/6 \\ 2/3 & -10/3 & 2/3 \\ 1/6 & 2/3 & 1/6 \end{bmatrix} \xleftarrow{1/3} h = \begin{bmatrix} 0.5 & 0 & 0.5 \\ 0 & -2 & 0 \\ 0.5 & 0 & 0.5 \end{bmatrix}$$

Impulse response
Matlab implementation



Frequency response

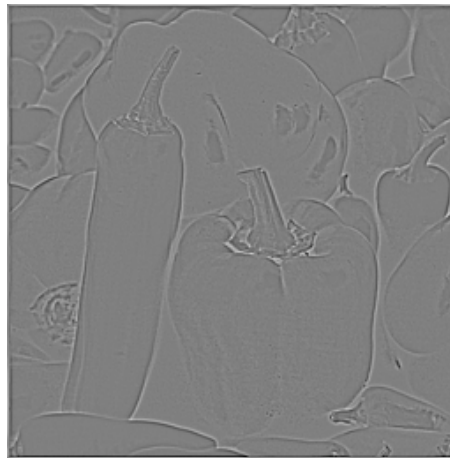
Laplacian approximation

4.2

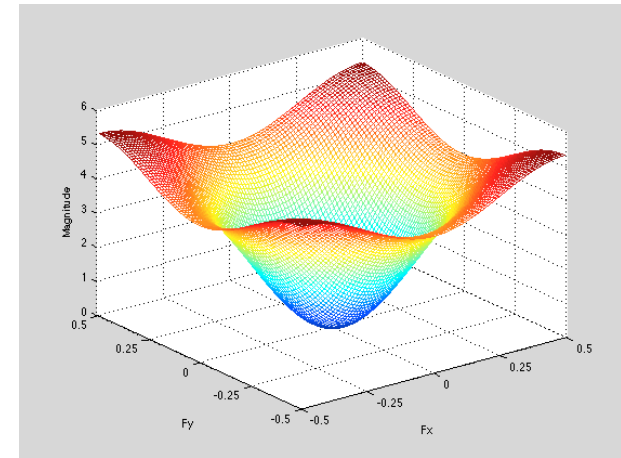
- The **Laplacian filter** detects contours estimating the zeros of the Laplacian function. It is not very robust against noise:
 - Typically, a filter to remove noise is applied previous to the Laplacian filter.
- It produces positive and negative values and, to visualize the result, a **contrast transform** is applied.



Original image



Laplacian image

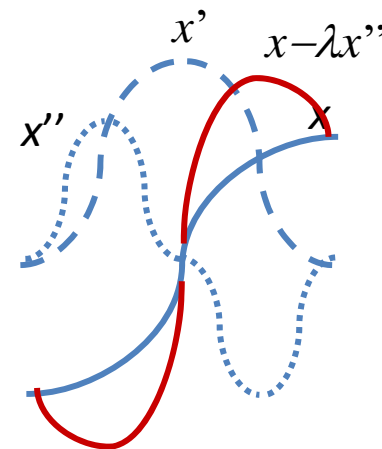


Frequency response

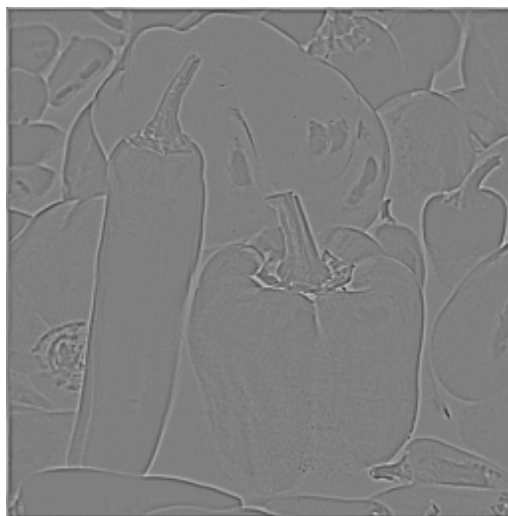
Application to enhancement

4.2

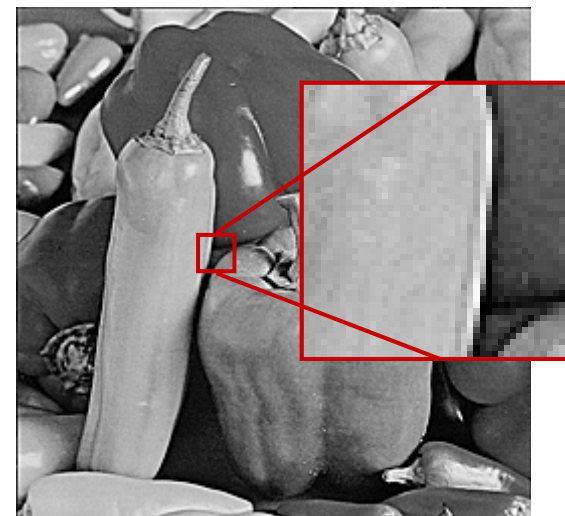
- For visualization purposes, it may be interesting to highlight the presence of contours in the image.
- This can be done by **unsharp masking**; that is, adding a weighted version of the Laplacian image to the original one.



Original image



Laplacian image



Enhanced image

Unit Structure

4.2

1. Introduction:

- Image convolution
- Image padding

2. Low-pass filters designed in the spatial domain:

- Example of application: Noise removal
- Trade-off noise removal vs blurring

3. High-pass filters designed in the spatial domain:

- Example of application: Contour detection

4. Filters designed in the frequency domain:

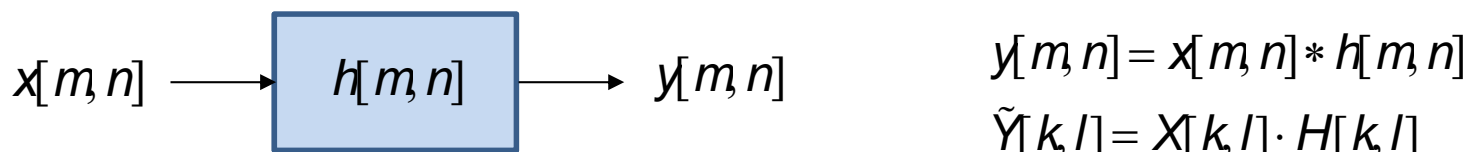
- Ringing effect

5. Summary and Conclusions

Ideal filters

4.2

Ideal filters (with **sharp transitions**) can be defined in the frequency domain.



- The filtering process implemented in the frequency domain implies:
 - Computing the DFT of $x[m, n]$ and center it: $X[k, l]$.
 - Filters are commonly defined centered.
 - Multiply the image transform by the filter: $\tilde{Y}[k, l] = X[k, l] \cdot H[k, l]$
 - Compute the inverse DFT of the resulting signal: $\tilde{Y}[k, l]$
 - Compute the real part of the result and de-center it again: $\tilde{y}[m, n]$
 - Although usually the filter has real values and, therefore, the filtered signal should be real, due to rounding effects final values may not be real.
- For visualization purposes, the image may need to be re-quantized.

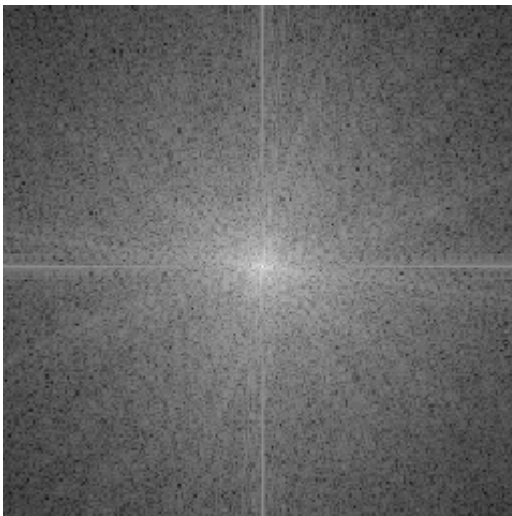
Ideal filters

4.2

Original
image



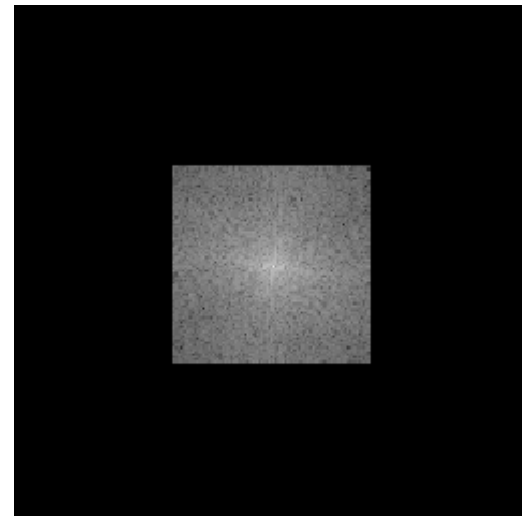
Transform
magnitude



Ideal low-pass
filtered image



Ideal low-pass filtered
transform magnitude

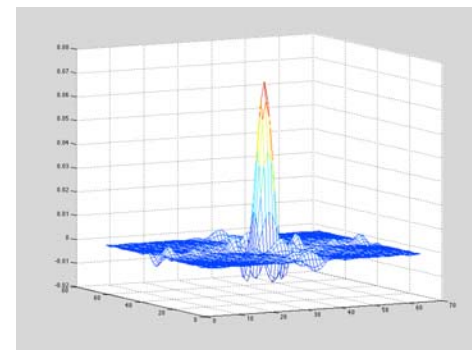


Ideal filters

4.2

- The ideal low-pass filtered image presents **ringing effect**.
- Ideal filters have as inverse transform **sinc functions**, whose lobes convolve with the transitions of the image producing these rings around contours.

Impulse response of ideal low-pass filter



Original image



Ideal low-pass filtered image



Unit Structure

4.2

1. Introduction:

- Image convolution
- Image padding

2. Low-pass filters designed in the spatial domain:

- Example of application: Noise removal
- Trade-off noise removal vs blurring

3. High-pass filters designed in the spatial domain:

- Example of application: Contour detection

4. Filters designed in the frequency domain:

- Ringing effect

5. Summary and Conclusions

Summary

4.2

- Image filters may be implemented in the **spatial domain** (typically, when the impulse response is short enough) or in the **frequency domain** (for long impulse responses or specific applications).
- Low-pass filters can be used for **noise removal**. They “average” the values within a neighborhood given by the impulse response of the filter. Filter values usually add up 1 to preserve the **image continuous component**.
- High-pass filters can be used for **contour detection**. They are designed as estimations of the **gradient** or the **Laplacian**. Gradient estimations approximate the derivative in the horizontal and vertical directions and, afterwards, combine these results into a single output.
- Ideal filters are designed in the frequency domain. Their sharp frequency transitions translates into the so-called **ringing effect** (which is the effect of the **spatial sinc function** convolving with the spatial transitions).