## Filtering Operation in Frequency Domain-1

**Fundamentals** 

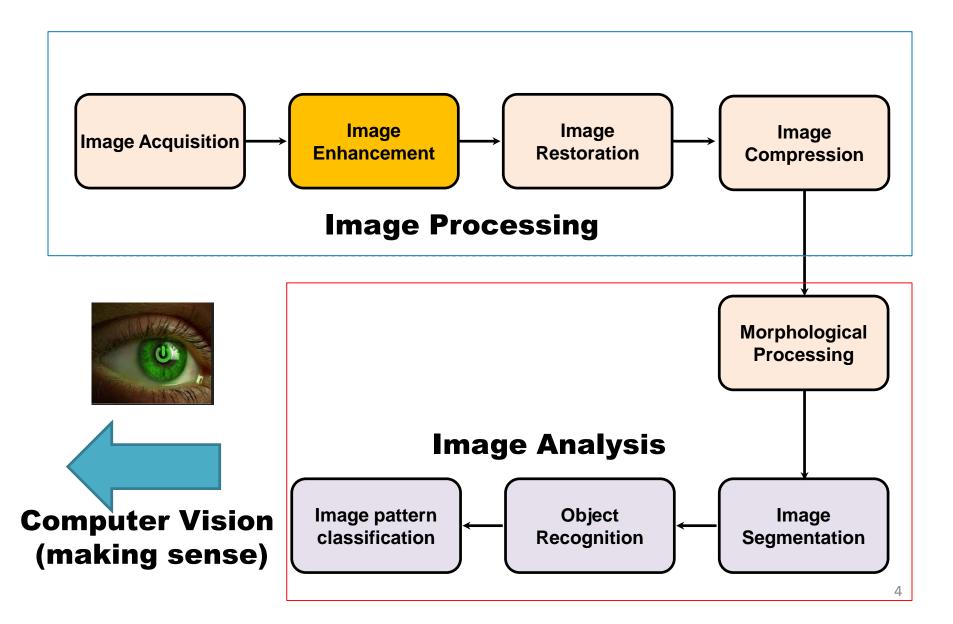
#### Recap

- DFT of one variable
- DFT of two variables
- How to overcome Wraparound Error?
- Properties of the 2-D DFT and IDFT

#### Lecture Objectives

- Filtering in Frequency Domain -Basic Observations
- Filtering in Frequency Domain Requirements
- What About the Padding for Filters in Frequency Domain?
- Steps for Filtering in the Frequency Domain
- Correspondence Between Filtering in Spatial and Frequency Domain
- Constructing Spatial Filters from Frequency Domain Filters
- Constructing Frequency Domain Filters from Spatial Filters

#### **Key Stages in DIP**



## Filtering in Frequency Domain Basic Observations

#### **Basic Observations**

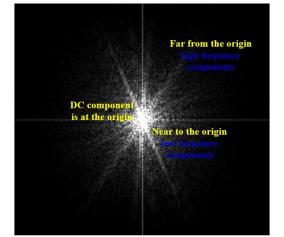
$$F(u,v) = \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x,y) e^{-j2\pi(ux/M + vy/N)}$$

for 
$$\mathbf{u} = 0, 1, 2, ..., M-1$$
 and  $\mathbf{v} = 0, 1, 2, ..., N-1$ 

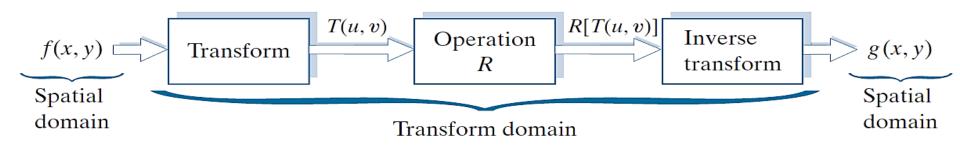
- Each term of F(u,v) contains all values of f(x,y), modified by multiplying the values of the exponential terms
  - Establishing a <u>direct one-to-one correspondence</u> between the components of image and its transform is not possible.
  - Only <u>general relationships</u> can be described like, the frequency range available in image.
- Frequency ≈ spatial rates of change in the image
  - Frequencies in the Fourier transform are intuitively related with patterns of intensity variations in the image.

#### Frequency Values with (u,v) Coordinates

- F(0,0) -> slowest varying frequency value (constant, 'dc' term)
  - Proportional to the average value of the entire image pixel intensities
- Values near the origin correspond to the slowly varying intensity components in image --> smoother regions in the image
  - Walls, clear sky, plain diffuse surfaces
- Values **far from the origin** correspond to the **faster varying intensity** components in image --> higher intensity variation regions in the image
  - edges



#### **Recall:** Image Domain Transforms



$$T(u,v) = \sum_{x=0}^{M-1} \sum_{y=0}^{N-1} f(x,y) \underbrace{r(x,y,u,v)}_{\text{Linear Transform}} \text{Linear Transform}$$

$$f(x,y) = \sum_{u=0}^{M-1} \sum_{v=0}^{N-1} T(u,v) \underbrace{s(x,y,u,v)}_{\text{Inverse transformation kernel}}$$
 Inverse Linear Transform

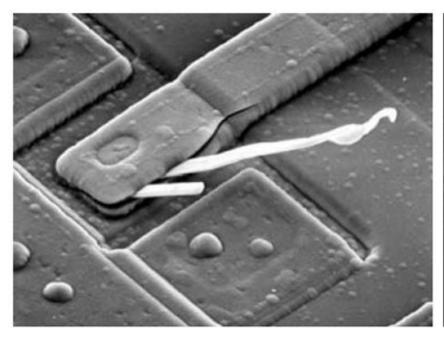
# Filtering in Frequency Domain Requirements

#### How to Filter in the Frequency Domain?

- Basic tool to work with: Fourier Transform.
- Spatial filtering works on the intensity values.
- Frequency domain filtering works on the Fourier transform values.
  - compute the Fourier transform of the image.
  - 2) modify the Fourier transform of an image (apply filters).
  - then compute the inverse transform to obtain the spatial domain representation of the processed result.
- What values do we have?
  - Fourier spectrum (magnitude)
  - Phase angle

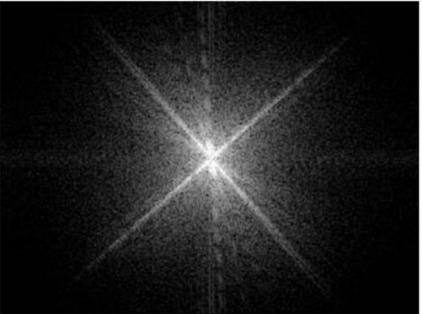
$$F(u,v) = |F(u,v)|e^{j\phi(u,v)}$$
$$|F(u,v)| = \left[R^2(u,v) + I^2(u,v)\right]^{1/2}$$
$$\phi(u,v) = \arctan\left[\frac{I(u,v)}{R(u,v)}\right]$$

#### Example



Scanning Electron Microscope image of an damaged Integrated Circuit

- Strong edges in ±45 ° directions
- The "defect" has edges in other directions
  - White, oxide protrusions



it's Fourier spectrum

- Strong edges in ±45° directions
- The "defect" is "visible" as patterns near the vertical direction

#### Filtering Operation in Frequency Domain

• Given (a padded) digital image, f(x, y), of size  $P \times Q$  pixels, the basic filtering equation in which we are interested has the form:

$$g(x,y) = \text{Real}\left\{\Im^{-1}\left[H(u,v)F(u,v)\right]\right\}$$

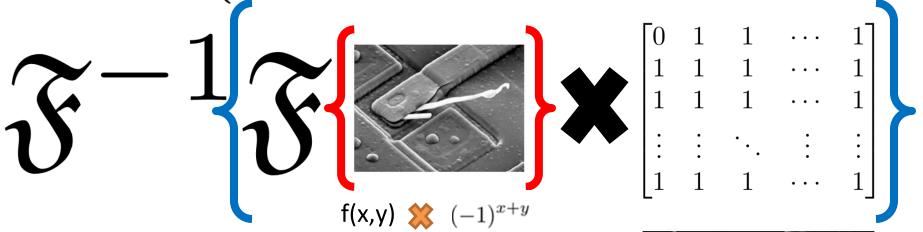
- $-\mathfrak{I}^{-1}$  is the IDFT
- F(u,v) is the **DFT** of the input image f(x, y) in which F(0,0) is **centered** at F(u,v) by multiplying the input image by  $(-1)^{x+y}$  prior to computing F(u,v)
- H(u,v) is a filter function which is symmetric about its center
- -g(x, y) is the filtered (output) image
- Functions F, H, and g are arrays of size  $P \times Q$ , the same as the padded input image
- The product H(u,v)F(u,v) is formed using elementwise multiplication

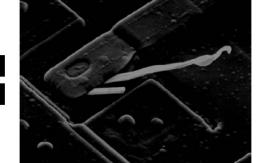
$$\begin{bmatrix} a_{11} & a_{12} \\ a_{21} & a_{22} \end{bmatrix} \odot \begin{bmatrix} b_{11} & b_{12} \\ b_{21} & b_{22} \end{bmatrix} = \begin{bmatrix} a_{11}b_{11} & a_{12}b_{12} \\ a_{21}b_{21} & a_{22}b_{22} \end{bmatrix}$$

#### Simple Frequency Domain Filter - example

To mask out the DC term (average intensity) - F(0,0)

$$H(u,v) = \begin{cases} 0 & \text{if } u = 0 \text{ and } v = 0 \\ 1 & \text{otherwise} \end{cases}$$

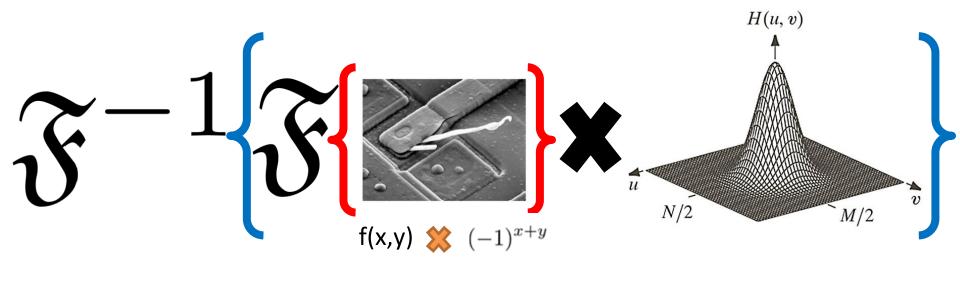




#### Frequency Domain Filter Nomenclature

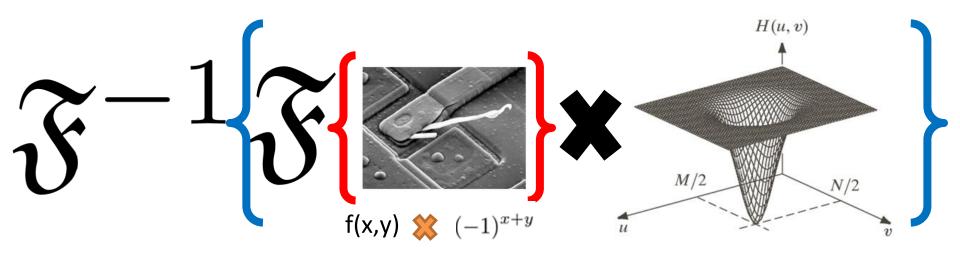
- Low frequencies: regions of smooth intensities
- High frequencies: abrupt transitions in the intensity values
- Lowpass filter: Pass the low frequencies, attenuate the high frequencies
  - Blur the image
  - Blur the edges
- Highpass filter: Pass the high frequencies, attenuate the low frequencies
  - Enhance edges, sharpen the image
  - Lower the contrast in the image

### Lowpass Filter



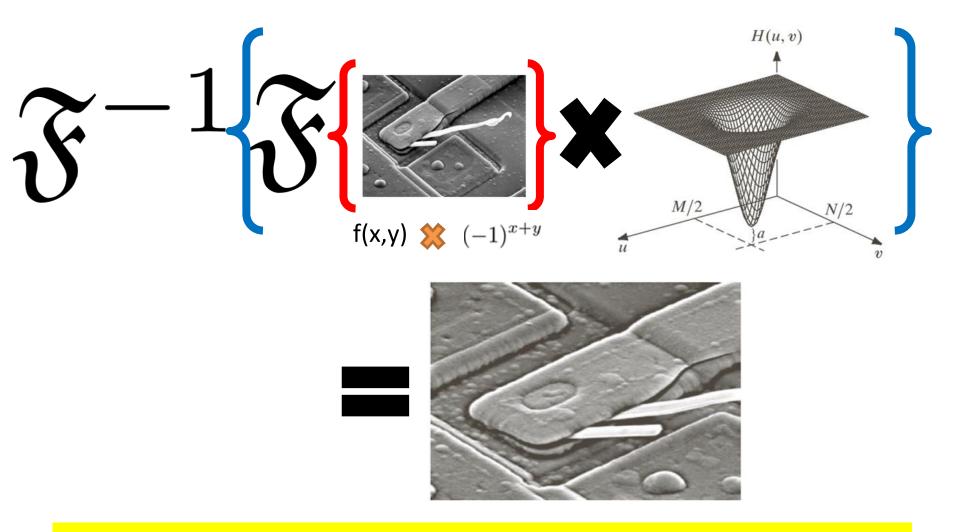


### Highpass Filter



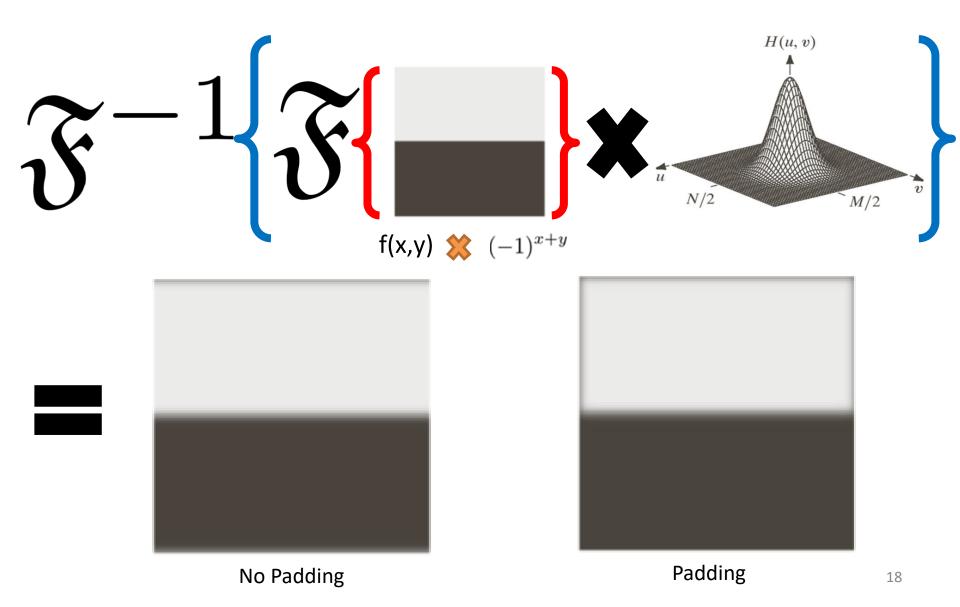


#### Offset Highpass Filter

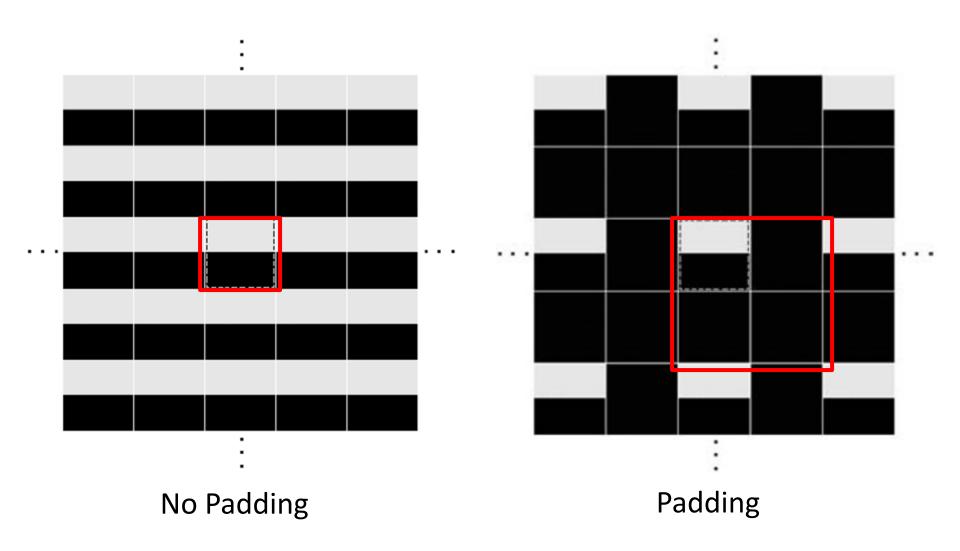


Adding a small constant to the highpass filter does not affect sharpening appreciably, but it does *prevent* elimination of the **dc term** and thus *preserves tonality*.

#### Filtering with Padding the input image



### Effect of Padding on Input Image



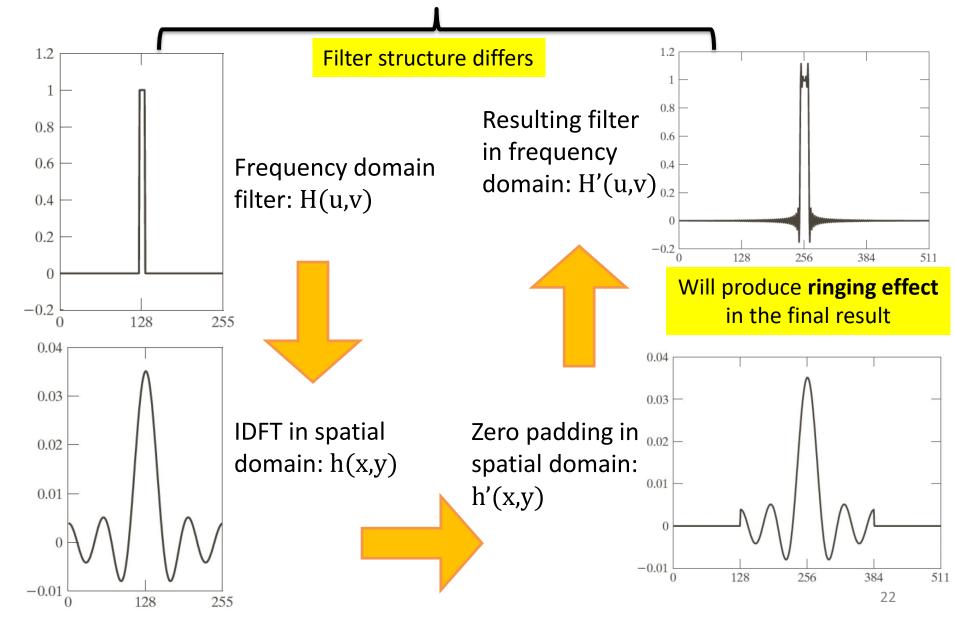
# What About the Padding for Filters in Frequency Domain?

#### Simple Idea (Not preferred)

#### **Steps**:

- 1. Construct the filter in the frequency domain having same size as the unpadded input image
- 2. Perform IDFT of this filter.
  - Frequency domain -> spatial domain
- 3. Do the Zero padding in the spatial domain.
- 4. Perform DFT to obtain the frequency domain filter.
  - Spatial domain -> frequency domain

#### Simple Idea (Not preferred)



#### Simple Idea (Not preferred) – Solution ??

#### **Steps**:

- 1. Pad the input image (in spatial domain of course...).
- Construct the filter in frequency domain having same size as the padded input image.
- This approach will result in wraparound error because no padding is used for the filter transfer function.
  - this error is mitigated significantly by the separation provided by padding the input image
  - it is preferable to ringing
  - Results in wraparound error, but smoother images

#### Effect of Frequency Filtering on the Phase Angle

$$F(u,v)=R(u,v)+jI(u,v)$$

- Substitute it in  $g(x,y)=F^{-1}[H(u,v)F(u,v)]$  $g(x,y)=F^{-1}[H(u,v)R(u,v)+jH(u,v)I(u,v)]$
- ☐ Phase angle is given by:

$$\phi(u,v) = \arctan\left[\frac{I(u,v)}{R(u,v)}\right]$$

- $\Box \Phi_{G}(u,v) = \Phi_{F}(u,v)$  since H(u,v) cancel out
- The filters that have no effect on the phase angle, are appropriately called as zero-phase-shift filters

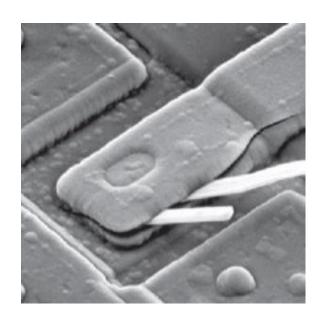
#### Effects of Phase Angle Change

- $F(u,v)=|F(u,v)|e^{j\phi(u,v)}$  Original image
- $F_1(u,v) = |F(u,v)|e^{j[-1]\phi(u,v)]} Phase angle \times (-1)$
- $F_2(u,v)=|F(u,v)|e^{j[0.25\phi(u,v)]}$  Phase angle × 0.25

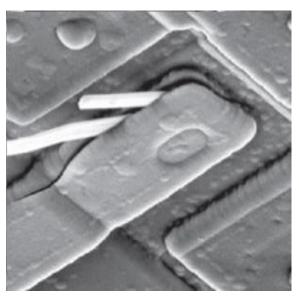
$$\mathfrak{F}^{-1}\left\{F(u,v)\right\} = \mathbb{F}_{1}(u,v)$$

$$\mathfrak{F}^{-1}\left\{F_{1}(u,v)\right\} = \mathbb{F}_{2}(u,v)$$

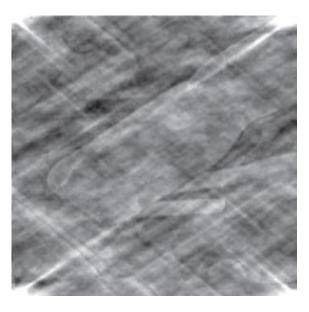
#### Effects of Phase Angle Change







Phase angle  $\times$  (-1)



Phase angle  $\times$  (0.25)

These two results illustrate the advantage of using <u>frequency-domain filters that</u> do not alter the phase angle.

# Steps for Filtering in the Frequency Domain

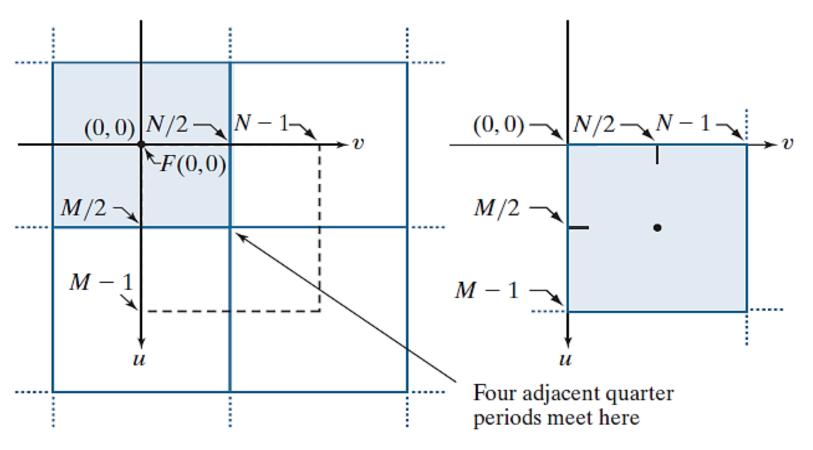
Important !!!

#### Steps for Filtering in the Frequency Domain

- 1. Given an input image f(x,y) of size  $M \times N$ , obtain the padding parameters P and Q (typically, P = 2M and Q = 2N).
- 2. Form a zero padded image  $f_p(x,y)$  of size  $P \times Q$  using zero-, mirror-, or replicate padding to the image f(x,y).

3. Multiply  $f_p(x,y)$  by  $(-1)^{x+y}$  to center its transform.

### 3. Multiply $f_p(x,y)$ by $(-1)^{x+y}$ to center its transform

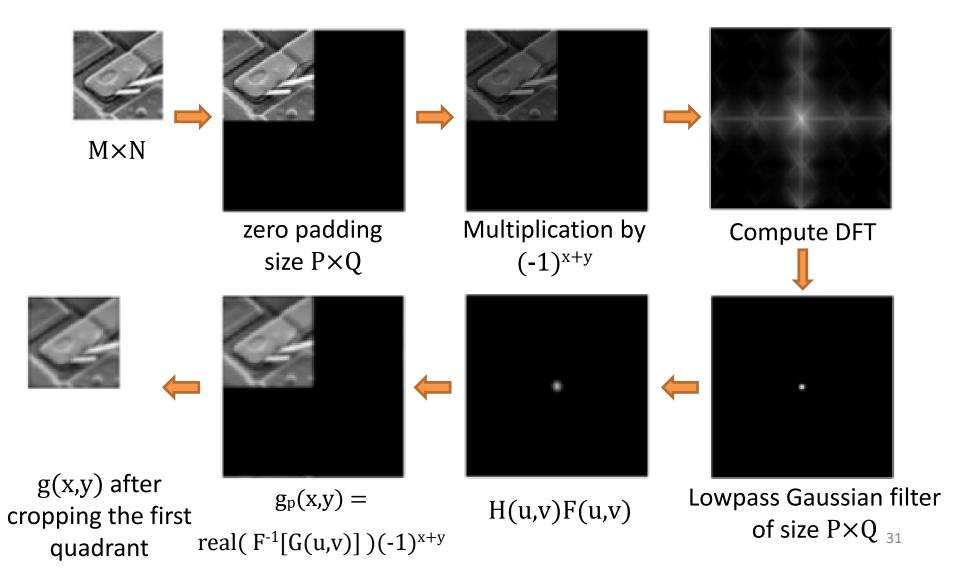


- $= M \times N$  data array computed by the DFT with f(x, y) as input
- $= M \times N$  data array computed by the DFT with  $f(x,y)(-1)^{x+y}$  as input
- ---- = Periods of the DFT

#### Steps for Filtering in the Frequency Domain

- 4. Compute the DFT, F(u,v) of the image  $f_p(x,y)$  from step-3.
- 5. Construct a real, symmetric filter H(u,v) of size  $P\times Q$  with center at the location (P/2,Q/2).
- 6. Form the product  $G(u,v)=H(u,v) \cdot F(u,v)$  using the elementwise multiplication operation for u=0,1,2,...,M-1 and v=0,1,2,...,N-1.
- 7. Obtain the filtered image (of size  $P \times Q$ ) by computing the IDFT of G(u,v):  $g_p(x,y) = \left( \operatorname{real} \left[ \Im^{-1} \left\{ G(u,v) \right\} \right] \right) (-1)^{x+y}$
- 8. Obtain the final image g(x,y) of the same size as the input image by extracting the  $M \times N$  region from the top, left quadrant of  $g_p(x,y)$ .

#### Filtering Explained with Example



### Correspondence Between Filtering in Spatial and Frequency Domain

## Correspondence between filtering in the Spatial and Frequency Domains

 The link between Spatial and Frequency Domains is the Convolution Theorem.

$$f \star h)(x, y) \Leftrightarrow (F \cdot H)(u, v)$$

• Filtering in frequency domain is an elementwise product of H(u,v) and F(u,v).

 Given: H(u,v), can we find corresponding h(x,y) filter in spatial domain?

#### Computing h(x,y)

- Let  $f(x,y) = \delta(x,y)$ , then F(u,v) = 1
- Hence,  $h(x,y) = \text{Real}\left\{\Im^{-1}\left[H(u,v)F(u,v)\right]\right\} = \Im^{-1}\left\{H(u,v)\right\}$ , and it is the inverse transform of the frequency domain filter.
- So, h(x,y) is the corresponding filter in the spatial domain.
- The converse is also true: i.e.,  $H(u,v)=F\{h(x,y)\}$
- We conclude that:  $h(x,y) \Leftrightarrow H(u,v)$  form a discrete Fourier transform pair for a filter.

#### Properties of h(x,y)

- h(x,y) is obtained by inverse discrete Fourier transform(IDFT) of the frequency domain filter with Fourier transform of an impulse function.
  - Also known as the impulse response of H(u,v)
- All quantities in discrete representations of H(u,v) and h(x,y) are finite.
  - Such filters are known as finite impulse response (FIR) filters
- Spatial convolution filtering is well suited (speed) for small kernels using hardware and/or firmware implementation.
- When working with general purpose computers, frequency-domain filtering using a **fast Fourier transform** (FFT) algorithm can be hundreds of times faster than using spatial convolution.

### Constructing Spatial Filters from Frequency Domain Filters

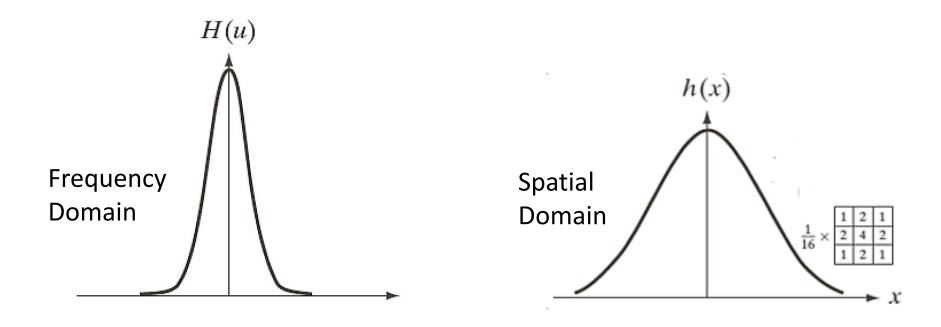
- Goal: use full size  $(P \times Q)$  frequency domain filters as a guide to specify the spatial filters for a much smaller neighborhood.
- We shall illustrate the method with Gaussian functions.
  - Recall: Both the forward and inverse Fourier transform of a Gaussian function are real Gaussian functions.

$$H(u) = Ae^{-u^2/2\sigma^2}$$
  $h(x) = \sqrt{2\pi\sigma}Ae^{-2\pi^2\sigma^2x^2}$ 

1-D Gaussian Fourier transform pair

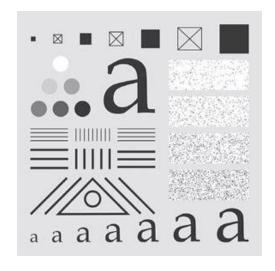
- Both H(u) and h(x) are real, and Gaussian functions
  - No complex terms (simplifies computations)
- Functions behave reciprocally. (sigma term)
  - − When H(u) has  $\sigma \rightarrow \infty => h(x)$  tends toward an impulse

#### **Lowpass Filter with a Gaussian Function:**



- All coefficients are positive.
- Largest value in center.
- Values decrease towards "outer" samples.

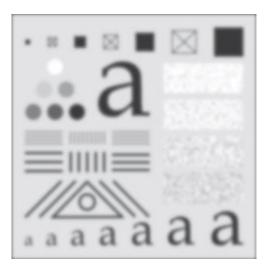
#### **Lowpass Filter with a Gaussian Function:**



Test image of size 1024 x 1024



Filtering with Gaussian kernel of size 21 x 21, with  $\sigma = 3.5$ 



Filtering with Gaussian kernel of size 43 x 43, with  $\sigma = 7$ 

#### **Highpass Filter with a Gaussian Function:**

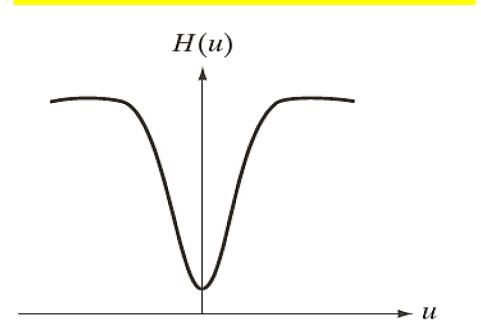
$$H(u) = Ae^{-u^2/2\sigma_1^2} - Be^{-u^2/2\sigma_2^2}$$
  

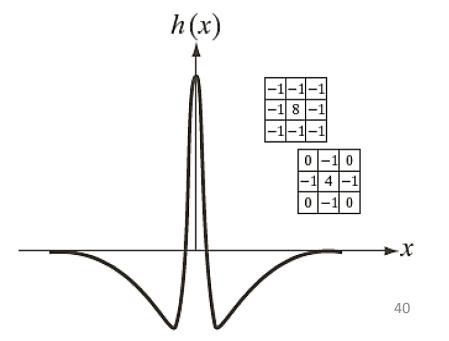
$$A \ge B \quad and \quad \sigma_1 \ge \sigma_2$$

Highpass Gaussian filter in **frequency domain** made of Two lowpass Gaussian
functions

$$h(x) = \sqrt{2\pi}\sigma_1 A e^{-2\pi^2 \sigma_1^2 x^2} - \sqrt{2\pi}\sigma_2 B e^{-2\pi^2 \sigma_2^2 x^2}$$

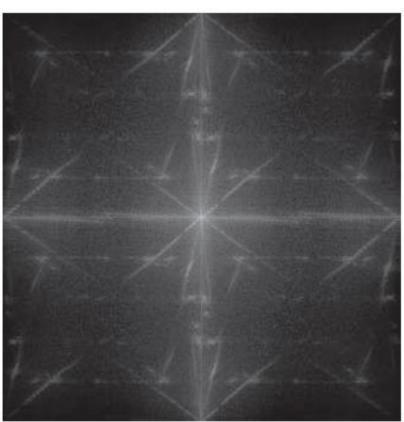
Corresponding Highpass Gaussian filter in spatial domain







f(x,y) of size  $600 \times 600$ 



Its F(u,v), centered

- Sobel Filter h(x,y)
  - $-3\times3$  kernel size
- Image Size f(x,y)
  - $-600\times600$
- To avoid wraparound error, We need zero padding for both f(x,) and h(x,y) !!!
  - $P \ge A + B 1$
  - $P \ge 600 + 3 1 = 602$

-1	0	1
-2	0	2
-1	0	1

Spatial Sobel filter h(x,y)

**Step-1:** Sobel function exhibits ODD symmetry. However, its 1<sup>st</sup> element is not zero.

**Step-2:** To make it an perfect **odd function**, we have to add to it a **leading row and column of 0's**.

$$h(x,y) = \begin{bmatrix} -1 & 0 & 1 \\ -2 & 0 & 2 \\ -1 & 0 & 1 \end{bmatrix} \longrightarrow \begin{bmatrix} 0 & 0 & 0 & 0 \\ 0 & -1 & 0 & 1 \\ 0 & -2 & 0 & 2 \\ \hline 0 & -1 & 0 & 1 \end{bmatrix}$$

For an **odd function**, when M is an **even number**, a **1-D odd sequence** has always **zero** values for the points at locations O and M/2.

 $\{0,-b,0,b\}$ 

**Step-3:** Place the Sobel filter from previous step into an larger array of zeros of the size  $602 \times 602$  with both of their **centers coincide**.

Centered at index (301,301)	•	•	•	•	•	•	•	•	•	•	•
	•	•	•	•	•	•	•	•	•	•	•
	•	•	•	0	0	0	0	•	•	•	•
	•	•	•	0	-1	0	1	•	•	•	•
		•		0	-2	0	2	•	•		
	•		•	0	-1	0	1	•	•	•	•
	•	•	•	•	•	•	•	•	•	•	•
	•	•	•	•	•	•	•	•	•	•	•
	•	•	•	•	•	•	•	•	•	•	
	•	•	•	•		•	•	•	•	•	

 $h(x,y) = 602 \times 602$  size array of zeros

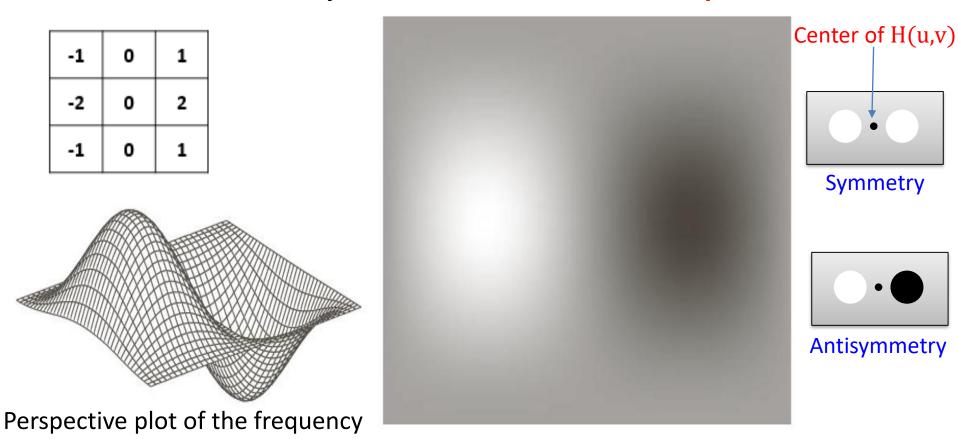
- <u>Very important point !!!</u> If odd (even) characteristic of a filter is not preserved, the filtering is not identical in spatial and frequency domains !!!.
- Step-4: Find the forward DFT of h(x,y) resulting in the required H(u,v) filter in the frequency domain.
  - \* Since h(x,y) is ODD and REAL function, H(u, v) will be purely imaginary.

#### Recall:

f(x,y) real and odd  $\Leftrightarrow F(u,v)$  imaginary and odd

#### The complete procedure to generate H(u,v) from h(x,y):

- 1. Pad h(x,y) to form a  $602\times602$  elements array:  $h_p(x,y)$ .
- 2. Multiply  $h_p(x,y)$  with  $(-1)^{x+y}$  to center the frequency domain filter.
- 3. Compute the forward DFT of the result in step (2): H(u,v).
- 4. Set the real part of H(u,v) to 0 to facilitate numerical precision or to eliminate outliers. H has to be purely imaginary.
- 5. Finally, multiply H(u,v) with  $(-1)^{u+v}$  to compensate for the shift to the center of the spatial filter in step (2).

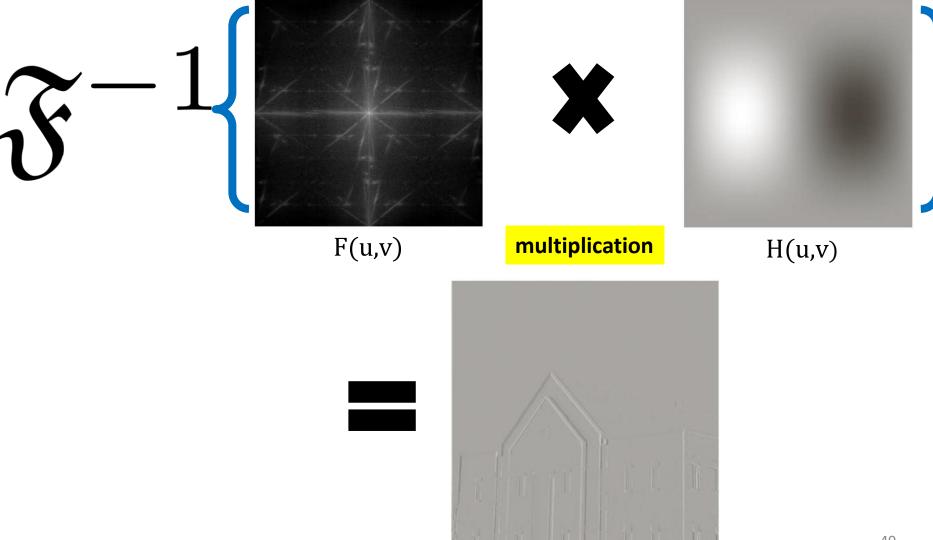


Note the **antisymmetry** in H(u,v) image about its center, a result of H(u,v) being odd.

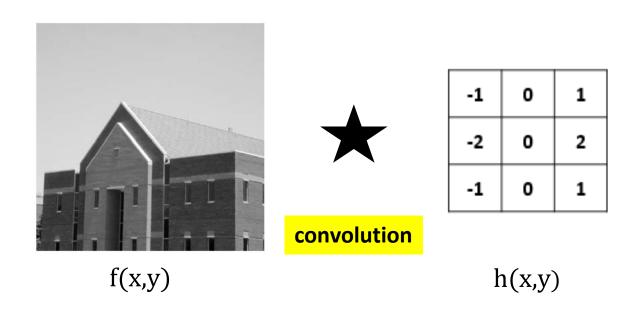
Fourier Spectrum of H(u,v)

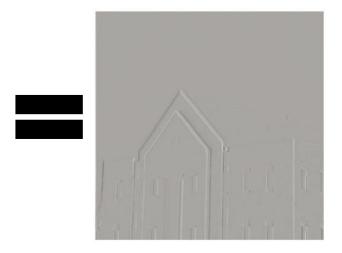
domain filter

#### Highpass filtering in Frequency Domain example

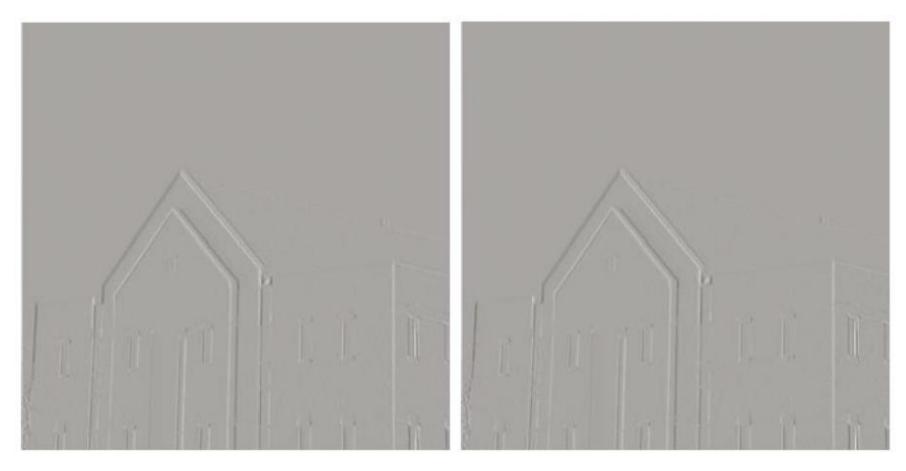


## Highpass filtering in Spatial Domain - example





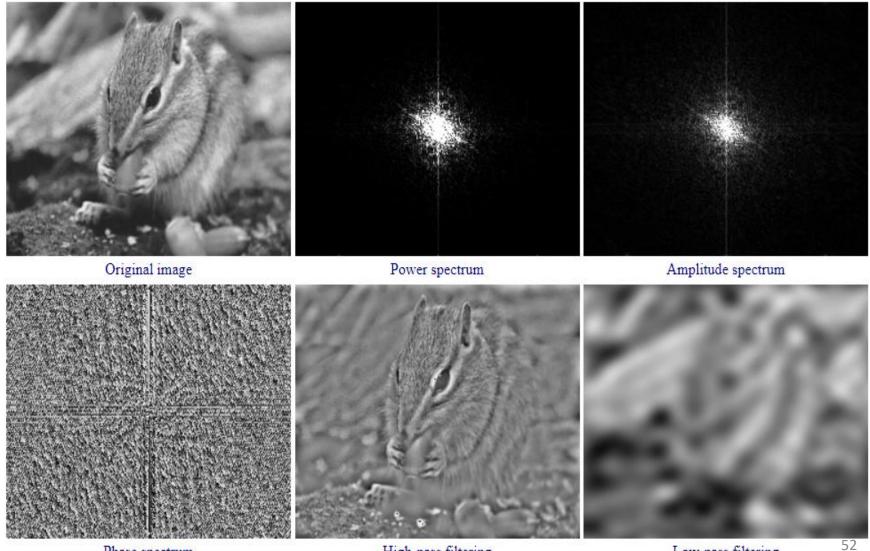
## Highpass filtering in Spatial Domain Vs. Frequency domain



Frequency Domain Filter Result

Spatial Domain Filter Result

#### Filtering - example



Phase spectrum High-pass filtering Low-pass filtering

#### Next Lecture

- Image Smoothing Using Lowpass Frequency Domain Filters
- Image Sharpening Using Highpass Frequency Domain Filters
- Laplacian in the Frequency Domain
- Homomorphic Filtering
- Selective Filtering