EE 4211 Computer Vision

Lecture 3B: Image enhancement (Frequency)

Semester B, 2021-2022

Spatial Domain vs. Frequency Domain

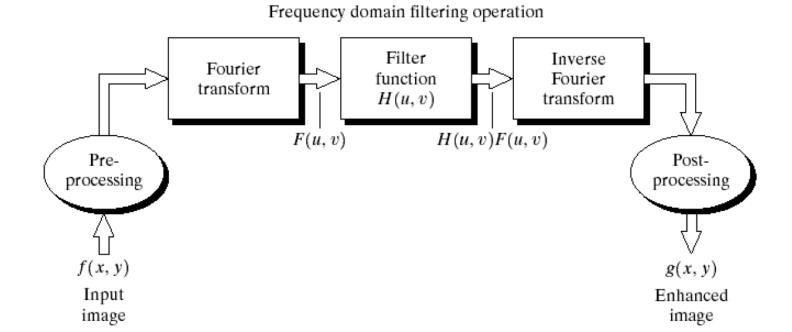
- Spatial Domain (image plane)
 - Techniques are based on direct manipulation of pixels in an image
- Frequency Domain
 - Techniques are based on modifying the spectral transform (in our course, we'll use Fourier transform) of an image
- There are some enhancement techniques based on various combinations of methods from these 2 domains

Lecture Outline

- Image Enhancement in Frequency Domain
- Filtering in Frequency Domain
 - Low-Pass Filtering
 - High-Pass Filtering
 - Laplacian Filtering
 - Homomorphic Filtering
 - Selective Filtering Bandpass/Bandreject, Notch Filters

- Images after transformation to frequency domain can be modified with frequency filters
- Nature of periodicity & conjugate symmetry.
- So, origin of spectrum is always shifted for
 - Display purpose
 - Filtering purpose

- To filter an image in the frequency domain:
 - Compute F(u,v) the DFT of the image
 - Multiply F(u,v) by a filter function H(u,v)
 - Compute the inverse DFT of the result



- Steps taken:
- Shift the origin of spectrum by multiplying image f(x,y) by (-1)^{x+y} before performing transformation to frequency domain.

$$f(x,y)(-1)^{(x+y)}$$
 $F\left(u-\frac{M}{2},v-\frac{N}{2}\right)$

G(u,v), the frequency spectrum obtained after applying frequency filter H(u,v) to frequency-transform image F(u,v) by multiplication, is defined as:

$$G(u,v) = H(u,v) * F(u,v)$$

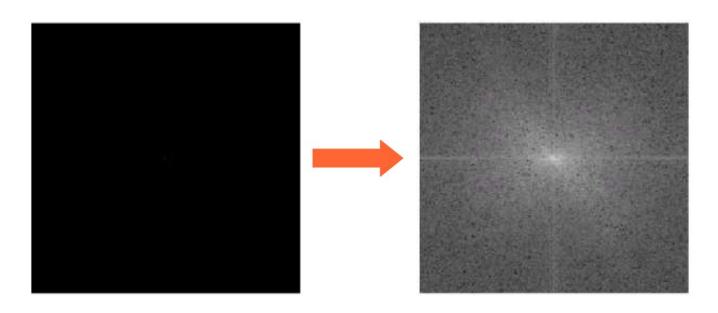
■ The filtered image, g(x,y) can then be obtained by performing the inverse transform to G(u,v):

$$g(x,y) = G^{-1}(u,v)$$

 Any shifting to origin performed before filtering should also be reversed after filtering.

- Steps taken:
- Enhance the visual information of the transformed image G(u,v) using log transform:

$$D(u,v) = k \log[1 + |G(u,v)|]$$



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Low-Pass Filter (in Freq. Domain)

- Lowpass filter remove high-frequency information, or allow low-frequency information to PASS through
- Useful for removing noise in images
- Also have undesired effect of blurring an image

Ideal Low-Pass Filter (ILPF)

- An ideal low-pass filter contain only 1's and 0's 1 for lower frequency and 0 for high frequency
- 2-D ideal lowpass filter (ILPF) is defined as:

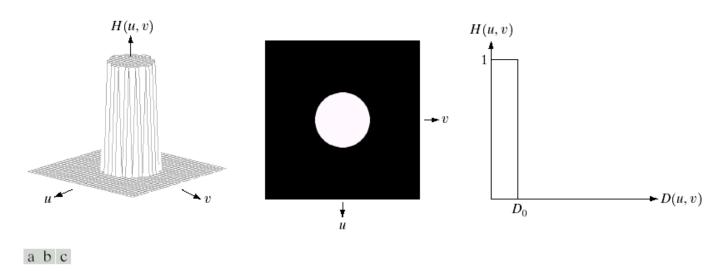
$$H(u, v) = \begin{cases} 1 & \text{if } D(u, v) \le D_0 \\ 0 & \text{if } D(u, v) > D_0 \end{cases}$$

• where D_0 is a positive constant and D(u,v) is the distance from point (u,v) to the origin (center) of the frequency rectangle. It is denoted as

$$D(u, v) = \sqrt{(u - M/2)^2 + (v - N/2)^2}$$

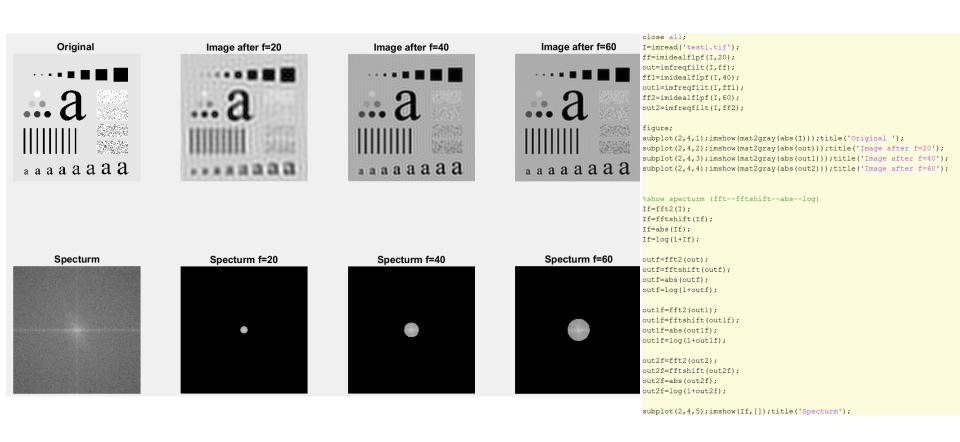
Ideal Low-Pass Filter (ILPF)

■ Ideal Low pass filter (ILPF): all frequencies inside a circle of radius D_0 are passed with no attenuation



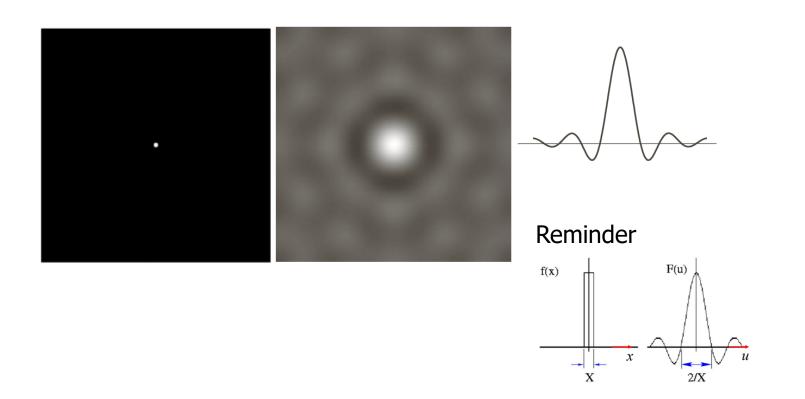
- (a)Perspective plot of an ideal low pass filter transfer function
- (b)Filter displayed as an image
- (c)Filter radial cross section

Images Filtered by ILPFs



Ideal Low-Pass Filter (ILPF)

- An ideal filter has undesired artifacts in images
- Presence of ripples/waves whenever there are boundaries in the image – "ringing effect"



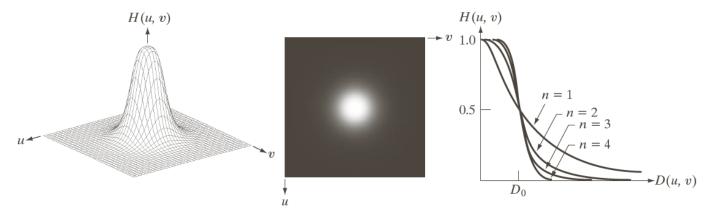
■ Transfer function of a Butterworth lowpass filter (BLPF) of order n, with D_0 cutoff frequency, is defined as

$$H(u, v) = \frac{1}{1 + [D(u, v)/D_o]^{2n}}$$

- where D(u,v) is the distance from origin of spectrum
- Butterworth lowpass filter Can specify order of filter, which determines steepness of slope in the transition of the filter function
 - Higher order of filter steeper slope closer to ideal filter

Advantages

- Reduces "ringing" while keeping clear cutoff
- Tradeoff between amount of ringing and sharpness of cutoff

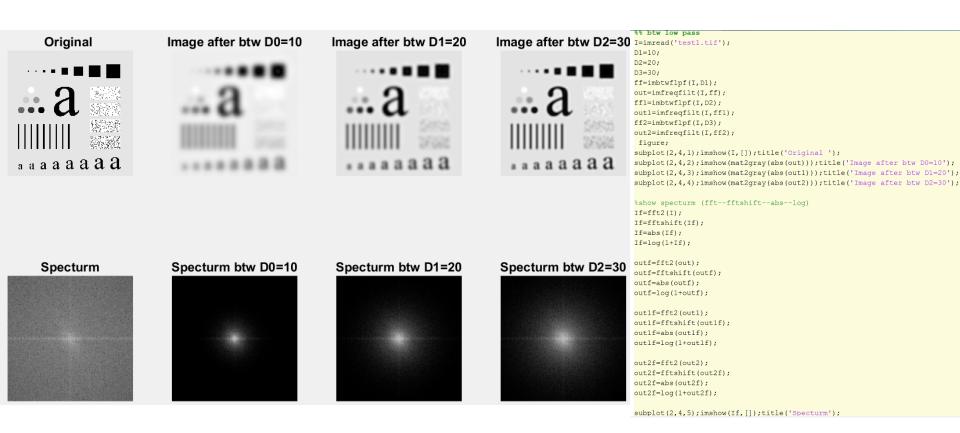


a b c

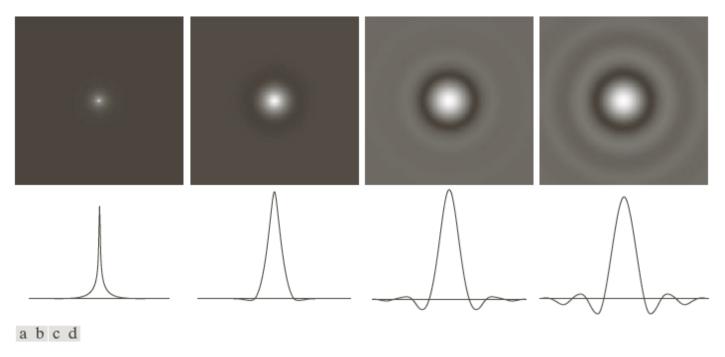
(a)Perspective plot of a Butterworth low pass filter transfer function

(b)Filter displayed as an image

(c) Filter radial cross section of orders 1 through 4



 Ringing properties increase if we increase the BLPF filter order, n



(a)-(d) Spatial representation of BLPFS of order 1, 2, 5, 20 and corresponding intensity profiles

Gaussian Low-Pass Filter (GLPF)

Gaussian lowpass filter in 2-D is defined as

$$H(u, v) = e^{-D^2(u, v)/2\sigma^2}$$

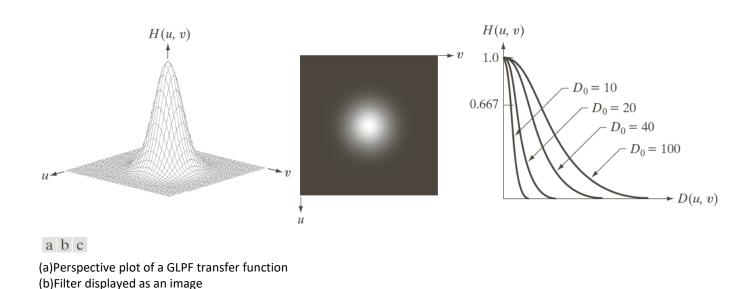
- where D(u,v) is the distance from the origin of spectrum, σ is the measure of spread of the Gaussian curve
- By letting $\sigma = D_0$, where D_0 is the cutoff frequency, we get

$$H(u,v) = e^{-D^2(u,v)/2D_0^2}$$

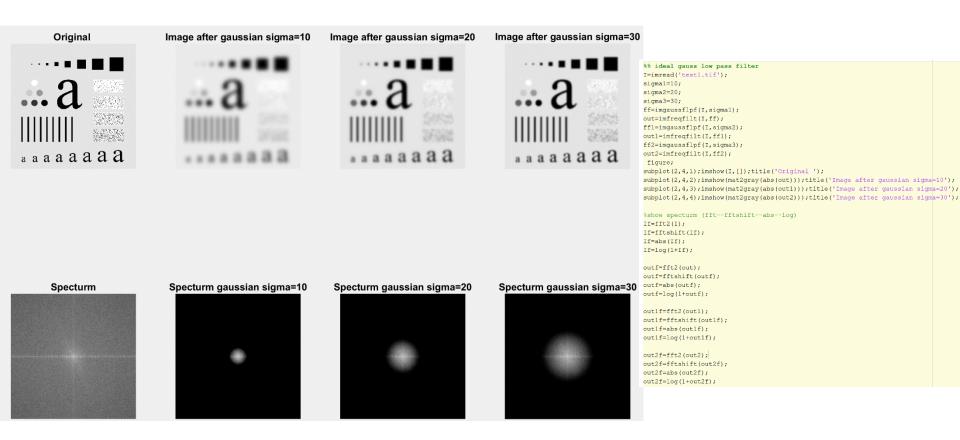
Gaussian Low-Pass Filter (GLPF)

- A Gaussian in the spatial domain also has the form of a Gaussian in the frequency domain
- No ringing, but allows high frequencies to pass

(c) Filter radial cross sections for various values of D₀



Images Filtered by GLPFs



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High-Pass Filter (in Freq. Domain)

- Highpass filter remove low-frequency information, or allow HIGH-frequency information to PASS through
- Useful for sharpening, edge enhancement
- Transfer function of the highpass filters can be obtained using the relation

$$H_{hp}(u, v) = 1 - H_{lp}(u, v)$$

where H_{lp}(u,v) is the transfer function of the corresponding lowpass filter

High-Pass Filters

Ideal highpass filter

$$H(u, v) = \begin{cases} 0 & \text{if } D(u, v) \le D_0 \\ 1 & \text{if } D(u, v) > D_0 \end{cases}$$

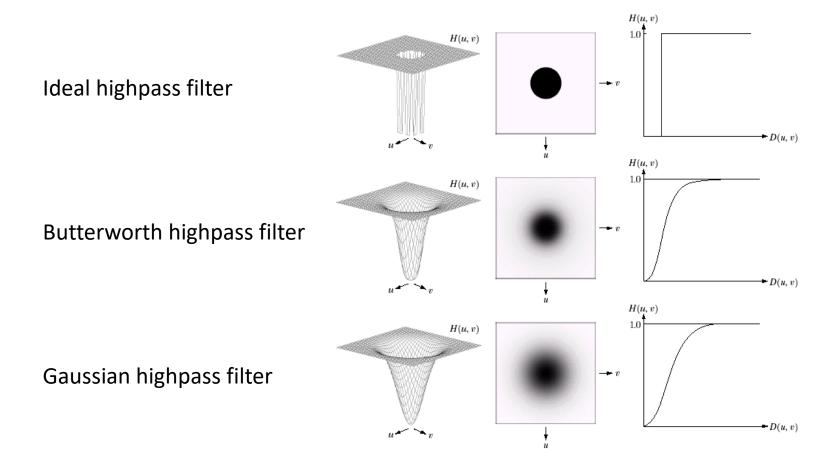
■ Butterworth highpass filter of order n and with cutoff frequency at a distance D_0 from the origin:

$$H(u, v) = 1 - \frac{1}{1 + [D_o/D(u, v)]^{2n}}$$

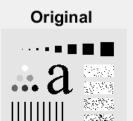
• Gaussian highpass filter with cutoff frequency at a distance D_0 from the origin

$$H(u,v) = 1 - e^{-D^2(u,v)/2D_0^2}$$

High-Pass Filters



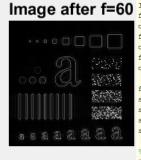
Filtering Results by IHPF



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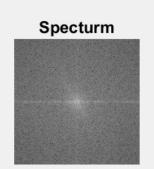


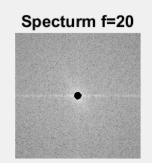


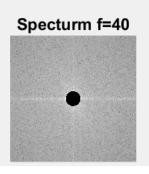


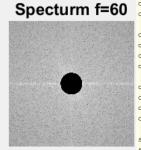
If=abs(If);
If=log(1+If);
outf=fft2(out);













Filtering Results by BHPF

Original

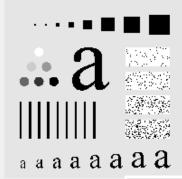


Image after btw f=10

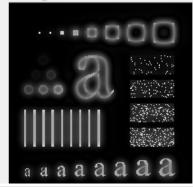


Image after btw f=20

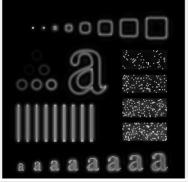


Image after btw f=30



```
%% btw high filter
I=imread('test1.tif');
sigma1=10;
sigma2=20;
sigma3=30;
ff=imbtwflpf(I, sigma1);ff=1-ff;
out=imfreqfilt(I,ff);
ff1=imbtwflpf(I, sigma2);ff1=1-ff1;
out1=imfreqfilt(I,ff1);
ff2=imbtwflpf(I, sigma3);ff2=1-ff2;
out2=imfreqfilt(I,ff2);
 figure;
subplot(1,4,1);imshow(I,[]);title('Original');
subplot(1,4,2);imshow(mat2gray(abs(out)));title('Image after btw f=10');
subplot(1,4,3);imshow(mat2gray(abs(out1)));title('Image after btw f=20');
subplot(1,4,4);imshow(mat2gray(abs(out2)));title('Image after btw f=30');
```

Filtering Results by GHPF

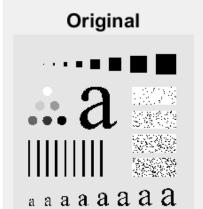


Image after sigma=10

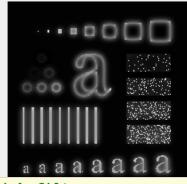


Image after sigma=20

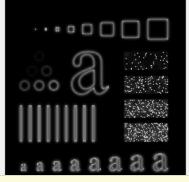
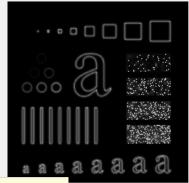


Image after sigma=30



```
%% gauss high filter
clc:
clear:
I=imread('test1.tif');
sigma1=10;
sigma2=20;
sigma3=30;
ff=imgaussflpf(I, sigma1);ff=1-ff;
out=imfreqfilt(I,ff);
ff1=imgaussflpf(I,sigma2);ff1=1-ff1;
out1=imfreqfilt(I,ff1);
ff2=imgaussflpf(I,sigma3);ff2=1-ff2;
out2=imfreqfilt(I,ff2);
figure;
subplot(1,4,1);imshow(I);title('Original');
subplot(1,4,2);imshow(mat2gray(abs(out)));title('Image after sigma=10');
subplot(1,4,3);imshow(mat2gray(abs(out1)));title('Image after sigma=20');
subplot(1,4,4);imshow(mat2gray(abs(out2)));title('Image after sigma=30');
```

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The Laplacian in Spatial Domain

The Laplacian is defined as follows:

$$\nabla^2 f = \frac{\partial^2 f}{\partial^2 x} + \frac{\partial^2 f}{\partial^2 y}$$

where the partial 2nd order derivative in the x direction is defined as follows:

$$\frac{\partial^2 f}{\partial^2 x} = f(x+1,y) + f(x-1,y) - 2f(x,y)$$

and in the y direction as follows:

$$\frac{\partial^2 f}{\partial^2 y} = f(x, y+1) + f(x, y-1) - 2f(x, y)$$

The Laplacian in Spatial Domain

So, the Laplacian can be given as follows:

$$\nabla^2 f = [f(x+1,y) + f(x-1,y) + f(x,y+1) + f(x,y+1) - 4f(x,y)]$$

We can easily build a filter based on this

0	1	0
1	-4	1
0	1	0

Laplacian Filter in Frequency Domain

$$\mathcal{F}\left[\frac{d^n f(x)}{dx^n}\right] = (ju)^n F(u)$$

The Fourier transform of Laplacian equation is

$$\mathcal{F}\left[\frac{\partial^2 f(x,y)}{\partial x^2} + \frac{\partial^2 f(x,y)}{\partial y^2}\right] = (ju)^2 F(u,v) + (jv)^2 F(u,v)$$
$$= -(u^2 + v^2) F(u,v)$$
$$\mathcal{F}[\nabla^2 f(x,y)] = -(u^2 + v^2) F(u,v)$$

This means that the Laplacian filter in frequency domain can be express as

$$H(u, v) = -(u^2 + v^2)$$

Basic preperties

Transform	x(t)	$\leftrightarrow X(u)$
scaling	x(at)	$\leftrightarrow \frac{1}{ a }X\left(\frac{u}{a}\right)$
shift	x(t- au)	$\leftrightarrow X(u) \mathrm{e}^{-j2\pi u \tau}$
modulation	$x(t)e^{j2\pi vt}$	$\leftrightarrow X(u-v)$
scale then shift	$x\left(\frac{t-\tau}{a}\right)$	$\leftrightarrow \frac{1}{ a } X\left(\frac{u}{a}\right) e^{-j2\pi u \tau}$
shift then scale	$x\left(\frac{t}{a}-b\right)$	$\leftrightarrow \frac{1}{ a } X\left(\frac{u}{a}\right) e^{-j2\pi uab}$
derivative	$\left(\frac{d}{dt}\right)^n x(t)$	$\leftrightarrow (j2\pi u)^n X(u)$
integral	$\int_{-\infty}^t x(\tau)d\tau$	$\leftrightarrow \frac{X(u)}{j2\pi u} + \frac{1}{2}X(0)\delta(u)$
conjugate	$x^*(t)$	$\leftrightarrow X^*(-u)$
transpose	x(-t)	$\leftrightarrow X(-u)$
inversion	$\int_{-\infty}^{\infty} X(u) e^{j2\pi ut} du$	$\leftrightarrow X(u)$
duality	X(t)	$\leftrightarrow x(-u)$
linearity	$ax_1(t) + bx_2(t)$	$\leftrightarrow aX_1(u) + bX_2(u)$
convolution	x(t) * h(t)	$\leftrightarrow X(u)H(u)$
correlation	$x(t) \star h(t)$	$\leftrightarrow X(u)H^*(u)$
real signals	if $x(t)$ is real	$\leftrightarrow X(u) = X^*(-u),$
	$\Rightarrow R(u) = R(-u),$	I(u) = -I(-u)
	X(u) = X(-u) ,	$\angle \{X(u)\} = -\angle \{X(-u)\}.$
causal signals	$x(t) = x(t)\mu(t)$	$\leftrightarrow X(u) = \frac{-j}{\pi u} * X(u)$
	$\Rightarrow I(u) = \frac{-1}{\pi u} * R(u),$	$R(u) = \frac{1}{\pi u} * I(u)$
Fourier series	$\sum_n c_n e^{j2\pi nt/T}$	$\leftrightarrow \sum_{n} c_{n} \delta\left(u - \frac{n}{T}\right)$
sampling theorem	$\sum_{n} x_n \operatorname{sinc}(2Bt - n)$	$\leftrightarrow \sum_n x_n \mathrm{e}^{-j\piu/B} \prod \left(\frac{u}{2B}\right)$

Laplacian Filter in Frequency Domain

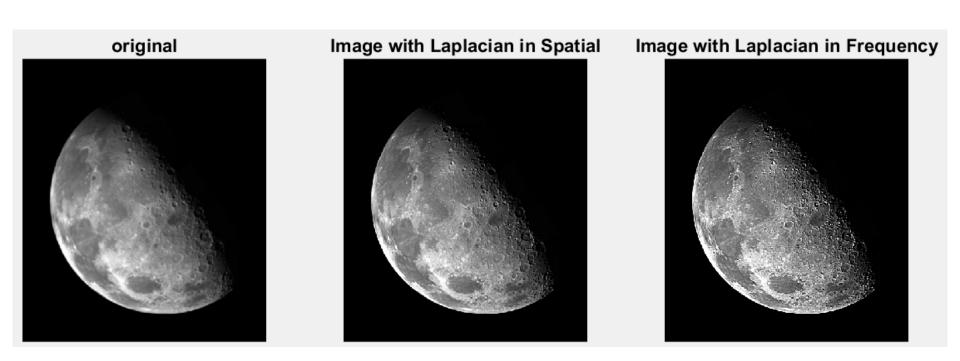
 As the filter origin was shifted to the center of image, we may shift the Laplacian filter in frequency domain by M/2 and N/2 respectively

$$H(u,v) = -[(u - M/2)^{2} + (v - N/2)^{2}]$$

Dual relationship in the familiar Fourier transform-pair notation:

$$\nabla^2 f(x,y) \Leftrightarrow -\left[(u - M/2)^2 + (v - N/2)^2 \right] F(u,v)$$

Example: Laplacian Filtering



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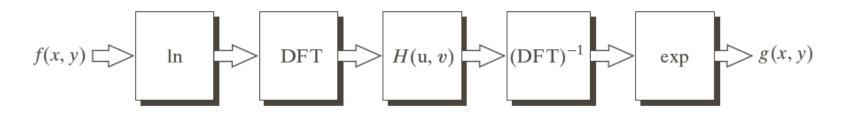
- Illumination-reflection Model:
 - Let f(x,y) be non-zero and finite image that $0 < f(x,y) < \infty$, f(x,y) may be characterized by components:
 - Source illumination (i(x,y)) incident on the scene being viewed
 - Amount of illumination reflection (r(x,y)) by the objects in the scene
- The two functions combine as a product to form

$$f(x,y) = i(x,y)*r(x,y)$$

where $0 < i(x,y) < \infty$,
and $0 < r(x,y) < 1$

- Illumination and Reflection have different characteristics:
 - Illumination components tend to be slow in spatial variation (low frequency components)
 - Reflection of various objects tends to vary abruptly (high frequency components)
- Better control can be achieved if the two components are separated by log function and filters are applied separately to each of the respective components.

The whole process can be summarized as follows



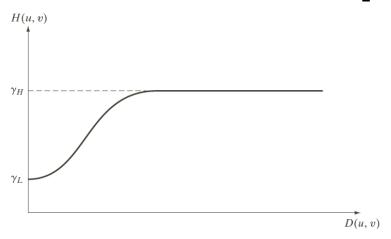
$$\begin{split} z(x, y) &= \ln(f(x, y)) = \ln(i(x, y)) + \ln(r(x, y)) \\ Z(u, v) &= \Im\{z(x, y)\} = \Im\{\ln(i(x, y))\} + \Im\{\ln(r(x, y))\} \\ &= F_i(u, v) + F_r(u, v) \\ S(u, v) &= H(u, v)Z(u, v) = H(u, v)F_i(u, v) + H(u, v)F_r(u, v) \\ s(x, y) &= \Im^{-1}\{H(u, v)F_i(u, v)\} + \Im^{-1}\{H(u, v)F_r(u, v)\} = i'(x, y) + r'(x, y) \\ g(x, y) &= e^{s(x, y)} = e^{i'(x, y)}e^{r'(x, y)} = i_0(x, y)r_0(x, y) \end{split}$$

- Filter H(u,v) can be designed such that it tends to decrease the contribution made by low frequencies (illumination, γ_L < 1) and amplify the contribution made by high frequencies (reflectance, $\gamma_H > 1$)
- The result is simultaneous dynamic range compression and contrast enhancement

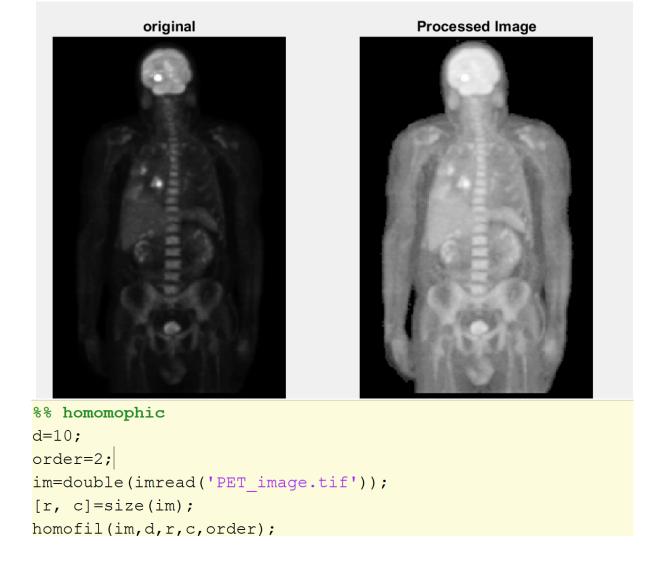
The curve shape can be approximated using the basic form of any of the ideal highpass filters. Example: using a slightlymodified form of GHPF gives:

$$H(u,v) = (\gamma_H - \gamma_L)[1 - e^{-cD^2(u,v)/D_0^2}] + \gamma_L$$

• where constant c controls the sharpness of the slope of the filter function as it transitions between γ_L and γ_H .



Example: Homomorphic Filtering



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Selective Filtering

- The previous filters all operate over the entire frequency rectangle. However, some applications only require processing of specific bands of frequencies or small regions of the frequency rectangle
 - Bandreject / Bandpass Filters: process specific bands
 - Notch Filters: process small regions of the frequency rectangle

Selective Filtering:

 Based on previous types of filters (Ideal, Butterworth, Gaussian), Bandreject filters can be constructed by adding an addition parameter – width of the band, W

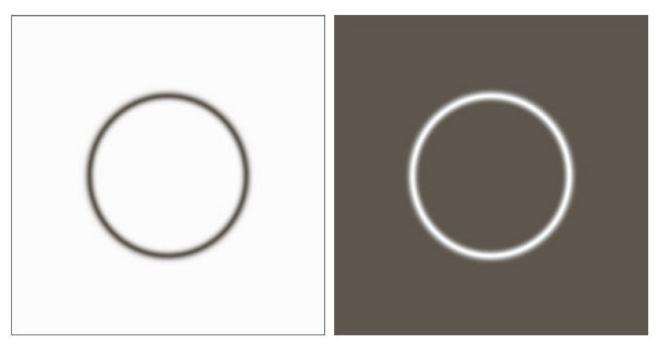
Bandreject filters. W is the width of the band, D is the distance D(u, v) from the center of the filter, D_0 is the cutoff frequency, and n is the order of the Butterworth filter. We show D instead of D(u, v) to simplify the notation in the table.

Ideal	Butterworth	Gaussian
$H(u, v) = \begin{cases} 0 & \text{if } D_0 - \frac{W}{2} \le D \le D_0 + \frac{W}{2} \\ 1 & \text{otherwise} \end{cases}$	$H(u, v) = \frac{1}{1 + \left[\frac{DW}{D^2 - D_0^2}\right]^{2n}}$	$H(u, v) = 1 - e^{-\left[\frac{D^2 - D_0^2}{DW}\right]^2}$

 Equivalent Bandpass filters can be obtained from a bandreject filter by inversing its effect

$$H_{BP}(u,v) = 1 - H_{BR}(u,v)$$

Selective Filtering



Bandreject Gaussian filter

Corresponding bandpass filter

Notch Filters

- A filter that rejects (or passes) specific frequencies
- Example: periodic noise corresponds to spikes or lines in the Fourier domain
- Can design a filter with zeros at those frequencies, this will remove the noise
- Examples:
 - Image mosaics
 - Scan line noise
 - Halftoning noise (moire patterns)

Steps in Notch Filtering

Look at spectrum |F(u,v)| of noisy image f(x,y), find frequencies corresponding to the noise

Create a mask image M(u,v) with notches (zeros) at those places, 1's elsewhere

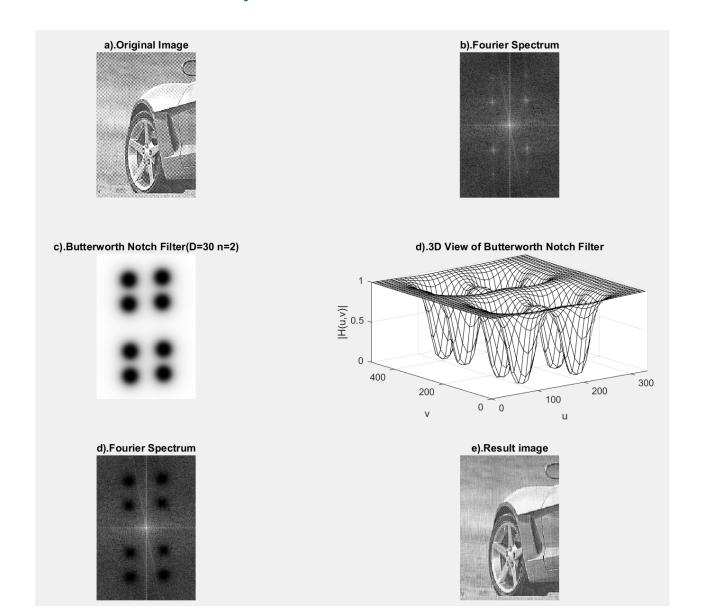
Multiply mask with original image transform; this zeros out noise frequencies

$$G(u,v) = M(u,v) F(u,v)$$

Take inverse Fourier transform to get restored image

$$g(x,y) = \mathcal{F}^{-1}(G(u,v))$$

Example Notch Filters



Summary

- An "ideal lowpass filter" passes all frequencies with magnitudes below a specific level, and attenuates all frequencies above that level.
- An "ideal highpass filter" does the opposite.
- A "notch" filter rejects (or passes) frequencies at a specific poin (the notch).

