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EE6310, Image and Video Processing

Linear Image Filtering

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Linear Image Filtering

- ▶ Wraparound and Linear Convolution
- ▶ Linear Image Filters
- ▶ Linear Image Denoising
- ▶ Linear Image Restoration
- ▶ Filter Banks

Linear Image Filtering – Illusions



Figure: Who is angry?

Linear Image Filtering – Illusions

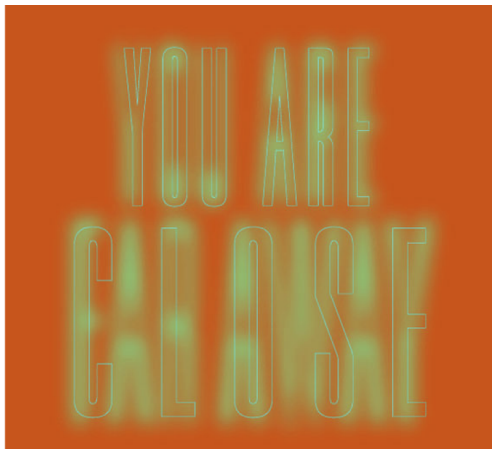


Figure: Close or far away?

Linear Image Filtering – Circular Convolution

- ▶ Modifying DFT of image changes its appearance - for e.g., applying a **zero-one mask**
- ▶ What happens if two image DFTs are **multiplied** pointwise?

$$\tilde{J} = \tilde{I}_1 \otimes \tilde{I}_2$$

- ▶ A very important question whose answer has profound consequences in image processing
- ▶ The inverse DFT of \tilde{J} is:

$$J(i, j) = \frac{1}{NM} \sum_{u=0}^{N-1} \sum_{v=0}^{M-1} \tilde{J}(u, v) W_N^{-ui} W_N^{-vj}$$

Linear Image Filtering – Circular Convolution

- Replacing $\tilde{J}(u, v)$ with $\tilde{l}_1(u, v) \otimes \tilde{l}_2(u, v)$

$$\begin{aligned} J(i, j) &= \frac{1}{NM} \sum_{u=0}^{N-1} \sum_{v=0}^{M-1} \tilde{l}_1(u, v) \otimes \tilde{l}_2(u, v) W_N^{-ui} W_N^{-vj} \\ &= \frac{1}{NM} \sum_{u=0}^{N-1} \sum_{v=0}^{M-1} \left\{ \sum_{n=0}^{N-1} \sum_{m=0}^{M-1} l_1(n, m) W_N^{un} W_M^{vm} \right\} \left\{ \sum_{p=0}^{N-1} \sum_{q=0}^{M-1} l_2(p, q) W_N^{up} W_M^{vq} \right\} W_N^{-ui} W_M^{-vj} \\ &= \frac{1}{NM} \sum_{n=0}^{N-1} \sum_{m=0}^{M-1} l_1(n, m) \sum_{p=0}^{N-1} \sum_{q=0}^{M-1} l_2(p, q) \sum_{u=0}^{N-1} \sum_{v=0}^{M-1} W_N^{u(n+p-i)} W_M^{v(m+q-j)} \\ &= \frac{1}{NM} \sum_{n=0}^{N-1} \sum_{m=0}^{M-1} l_1(n, m) \sum_{p=0}^{N-1} \sum_{q=0}^{M-1} l_2(p, q) \cdot NM \cdot \delta(n+p-i, m+q-j) \\ &= \sum_{n=0}^{N-1} \sum_{m=0}^{M-1} l_1(n, m) l_2[(i-n)_N, (j-m)_M] \\ &= \sum_{p=0}^{N-1} \sum_{q=0}^{M-1} l_1[(i-p)_N, (j-q)_M] l_2(p, q) = l_1 \# l_2 \end{aligned}$$

Linear Image Filtering – Circular Convolution

- ▶ $p_N = p \bmod N$
- ▶ $I_1 \# I_2$ is the **circular convolution** of I_1 with I_2
- ▶ Like **linear convolution** except with indices taken **modulo N**, **M**

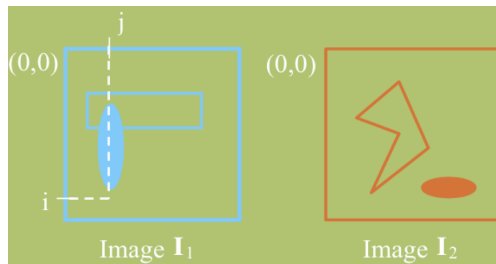


Figure: Depicting circular convolution

Linear Image Filtering – Circular Convolution

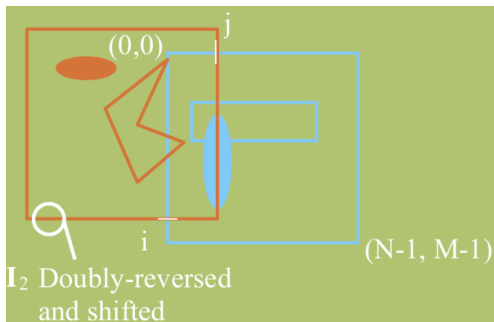


Figure: Without circular convolution

- Modulo arithmetic defines the product for all $0 \leq i \leq N - 1; 0 \leq j \leq M - 1$

Linear Image Filtering – Computing Circular Convolution

- ▶ Direct computation of $I_1 \# I_2$ is simple but expensive
- ▶ For an $N \times M$ image:
 - ▶ For each of NM coordinates: NM multiplications and additions
 - ▶ For the entire image: $NM.NM$ multiplications
 - ▶ If $N = M = 512$, it turns out $2^{36} = 6.9 \times 10^{10}$
- ▶ An alternative is to use FFT to compute $\#$
- ▶ $J = I_1 \# I_2 = \text{IFFT}_{N \times M}[\text{FFT}_{N \times M}[I_1] \otimes \text{FFT}_{N \times M}[I_2]]$
- ▶ Computing an $N \times M$ FFT is $O[NM.\log(NM)]$, and so is the computation of $\#$
- ▶ To make $\#$ useful, it must be modified

Linear Image Filtering – Linear Convolution

- ▶ Cyclic convolution is a consequence of the **DFT**, which is a sampled **DSFT**.
- ▶ If two **DSFTs** are multiplied together, then useful **linear convolution** results:
$$\tilde{J}_D = \tilde{I}_{D1} \tilde{I}_{D2} \implies J(i,j) = I_1(i,j) * I_2(i,j)$$
- ▶ **Cyclic convolution** is an **artifact** of sampling the DSFT – which causes **spatial periodicity**
- ▶ Most of circuit theory, optics and analog filter theory based on **linear convolution**

Linear Image Filtering – Linear Convolution

- ▶ Linear digital filter theory also requires the concept of **linear convolution**
- ▶ It turns out that **circular convolution** can be used to **compute** linear convolution
- ▶ However, circular convolution has drawbacks and cannot be used as-is
- ▶ Consider the average filter: output is the average of pixel values in a square window

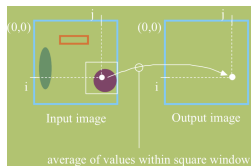


Figure: Averaging filter

Linear Image Filtering – Averaging Filter

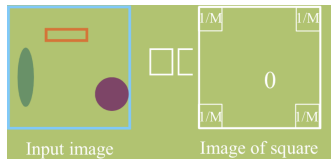


Figure: Averaging using circular convolution

The averaging filter may be expressed as the circular convolution of the image with the image of a square with intensity $1/M$ where M is the number of pixels in the square.

Linear Image Filtering – Averaging Filter

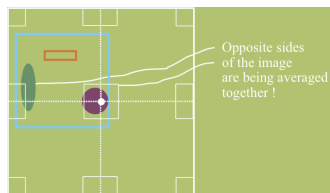


Figure: Undesirability of circular convolution

- ▶ Wraparound effects occur at image borders
- ▶ Desirable to average only neighboring elements and weight images based on **spatial ordering** rather than DFT-induced **periodic ordering**
- ▶ Large filters **worsen** effect

Linear Image Filtering – Linear Convolution by Zero Padding

- ▶ **Conceptually simple:** Pad images to be convolved with zeros
- ▶ **Typically**, both image arrays are doubled in size
- ▶ **Wraparound** eliminated since the “moving” image is weighted by zero outside image domain

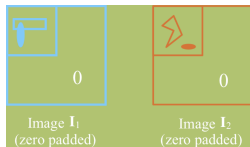


Figure: Zero padded images

Linear Image Filtering – Visualizing Wraparound Cancellation

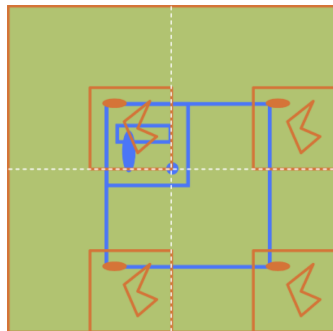


Figure: Wraparound cancellation

Remember, summation takes place within blue region
($0 \leq i \leq 2N - 1; 0 \leq j \leq 2M - 1$)

Linear Image Filtering – DFT Computation of Linear Convolution

- ▶ Let J', I'_1, I'_2 be **zero-padded** $2N \times 2M$ versions of $J = I_1 * I_2$
- ▶ If $J' = I'_1 \# I'_2 = \text{IFFT}_{2N \times 2M}[\text{FFT}_{2N \times 2M}[I'_1] \text{FFT}_{2N \times 2M}[I'_2]]$, the $N \times M$ image with elements $J(i, j) = J'(i, i); 0 \leq i \leq N - 1, 0 \leq j \leq M - 1$ contains the **linear convolution** result
- ▶ Therefore, by multiplying **zero-padded** DFTs and taking the IFFT, we get $J = I_1 * I_2$
- ▶ The linear convolution is larger than $N \times M$ ($2N \times 2M$) but the interesting part is contained in $N \times M$ region of J
- ▶ To convolve an $N \times M$ image with a $P \times Q$ filter:
 - ▶ If $P, Q < N, M$, pad filter with zeros to size $N \times M$
 - ▶ If $P, Q \ll N, M$, faster to convolve in space domain

Linear Image Filtering – Direct Linear Convolution

- ▶ Assume l_1 and l_2 are not periodically extended and assume $l_1(i, j) = l_2(i, j) = 0$ if $i < 0, j < 0$ and $i > N - 1, j > M - 1$
- ▶ In this case,

$$J(i, j) = l_1(i, j) * l_2(i, j) = \sum_{n=0}^{N-1} \sum_{m=0}^{M-1} l_1(n, m) l_2(i - n, j - m)$$

Linear Image Filtering – Goals

- ▶ Linear filtering refers to a process that transforms an image / by linear convolution
- ▶ Goals include transforming an image:
 - ▶ Into one of **better quality**
 - ▶ With certain features **enhanced**
 - ▶ With certain features **de-emphasised** or **removed**
- ▶ Specifically:
 - ▶ **Smoothing:** remove noise due to bit errors, transmission etc
 - ▶ **Deblurring:** increase **sharpness** of blurred images
 - ▶ **Sharpening:** emphasize significant features like **edges**
 - ▶ **Combinations** of these

Linear Image Filtering – Characterizing Linear Filters

- ▶ Any **linear** filter can be characterized in one of two equivalent ways:
 - ▶ The **impulse response** $\mathbf{H} = H(i, j)$
 - ▶ The **frequency response** $\tilde{\mathbf{H}} = \tilde{H}(u, v)$
- ▶ $\tilde{\mathbf{H}} = \text{DFT}[\mathbf{H}]$, $\mathbf{H} = \text{IDFT}[\tilde{\mathbf{H}}]$
- ▶ The **frequency response** describes how image frequencies are affected by the system
- ▶ Since $\tilde{H}(u, v) = |\tilde{H}(u, v)| \exp\{\sqrt{-1} \angle \tilde{H}(u, v)\}$, an image frequency component at $(u, v) = (a, b)$ is amplified or attenuated by $|\tilde{H}(a, b)|$ and shifted by $\angle \tilde{H}(a, b)$

Linear Image Filtering – Frequency Response Example

- ▶ If the **input** to a system H is a cosine image:

$$I(i, j) = \cos[2\pi(\frac{b}{N}i + \frac{c}{M}j)] = \frac{1}{2}\{W_N^{bi}W_M^{cj} + W_N^{-bi}W_M^{-cj}\}$$

- ▶ The **output** is:

$$J(i, j) = I(i, j) * H(i, j) = |\tilde{H}(b, c)| \cos[2\pi(\frac{b}{N}i + \frac{c}{M}j) + \angle \tilde{H}(b, c)]$$

- ▶ As with 1-D signals, the impulse response is an effective way to model responses since:

$$I(i, j) = \sum_{n=0}^{N-1} \sum_{m=0}^{M-1} I(n, m) \delta(i - n, j - m)$$

Linear Image Filtering – Linear Filter Design

- ▶ Given **analog** spec: $H_c(x, y) \xleftrightarrow{\mathfrak{F}} \tilde{H}_c(\Omega, \Lambda)$
- ▶ Two **simple** methods of filter design:
 - ▶ Space-sampled approximation:
 $(X = 1, Y = 1), H(i, j) = H_C(i, j); -\infty < i, j < \infty$
 - ▶ Frequency-sampled approximation using DFT:

$$\tilde{H}(u, v) = \sum_{n=-\infty}^{\infty} \sum_{m=-\infty}^{\infty} \tilde{H}_c[(\frac{u}{N} - n), (\frac{v}{M} - m)]$$

Linear Image Filtering – Space-Sampled Approximation

- ▶ Truncate the continuous response:
 $H_{trunc}(i, j) = H_C(i, j); 0 \leq |i| \leq (N/2)-1; 0 \leq |j| \leq (M/2)-1$
- ▶ Truncation of the analog spec leads to Gibbs phenomena at jump discontinuities of $\tilde{H}_C(\Omega, \Lambda)$
- ▶ DFT/IDFT pair $\tilde{H}_{trunc}(u, v) \xLeftrightarrow{\text{IDFT}} H_{trunc}(i, j)$

Linear Image Filtering – Frequency Sampled Approximation

- ▶ $\tilde{H}_{fs}(u, v) = \tilde{H}_C(\frac{u}{N}, \frac{v}{M}); 0 \leq |u| \leq (N/2) - 1, 0 \leq |v| \leq (M/2) - 1$
- ▶ DSFT NOT specified between samples
- ▶ CFT is **centered** and **non-periodic**
- ▶ The discrete impulse response is then: $H_{fs}(i, j) \xLeftrightarrow{\text{DFT}} \tilde{H}_{fs}(u, v)$

Linear Image Filtering – Low-Pass, Band-Pass and High-Pass Filters

- ▶ **Low-pass:** Attenuates all but “lower” frequencies
 - ▶ **smooth** noise
 - ▶ **blur** detail to highlight gross features
- ▶ **Band-pass:** Attenuates all but “middle” frequencies
 - ▶ **Special-purpose**
- ▶ **High-pass:** Attenuates all but “higher” frequencies
 - ▶ **enhance** image detail and contrast
 - ▶ **remove** image blur

Linear Image Filtering – Example Low-Pass Filter

- ▶ To smooth an image: replace each pixel in a noisy image by the **average** of its $M \times M$ neighbors
- ▶ How to pick window size M ?
- ▶ Example: for a 512×512 image, **typical** window sizes range from 3×3 , 5×5 , ..., 15×15 (lots of smoothing)

Linear Image Filtering – Example Low-Pass Filter

- Design an **ideal low-pass filter** in the DFT domain:

$$\tilde{H}(u, v) = \begin{cases} 1, & \text{if } \sqrt{u^2 + v^2} \leq U_{cutoff} \\ 0, & \text{otherwise} \end{cases}$$

- Possibly useful if **radial** frequency U_{cutoff} can be estimated in the original image

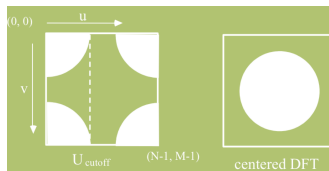


Figure: Ideal low-pass filter

Linear Image Filtering – Example Low-Pass Filter

- ▶ The **gaussian filter** with frequency response $\tilde{H}_C(\Omega, \Lambda) = \exp[-2(\pi\sigma)^2(\Omega^2 + \Lambda^2)]$, hence $\tilde{H}(u, v) = \exp\{-2\pi^2\sigma^2[(\frac{u}{N})^2 + (\frac{v}{M})^2]\}$
- ▶ Response falls quickly at high frequencies
- ▶ **Important** low-pass filter

Linear Image Filtering – Example Band-Pass Filter

- ▶ Define a BP filter as the **difference of two LPFs** that differ only by a scaling factor
- ▶ Common example is a **difference of gaussian (DOG)** filter:
$$\tilde{H}_C(\Omega, \Lambda) = \exp[-2(\pi\sigma)^2(\Omega^2 + \Lambda^2)] - \exp[-2(K\pi\sigma)^2(\Omega^2 + \Lambda^2)]$$

hence $\tilde{H}(u, v) =$
$$\exp\{-2\pi^2\sigma^2[(\frac{u}{N})^2 + (\frac{v}{M})^2]\} - \exp\{-2K^2\pi^2\sigma^2[(\frac{u}{N})^2 + (\frac{v}{M})^2]\}$$
- ▶ Typically, $K \approx 1.5$
- ▶ DOG filters are very useful for image analysis and human visual modeling

Linear Image Filtering – Example High-Pass Filter

- ▶ **Laplacian** filter is important as well
- ▶ A **severely truncated** approximation is
 $\tilde{H}_C(\Omega, \Lambda) = A(\Omega^2 + \Lambda^2)$, hence $\tilde{H}(u, v) = A[(\frac{u}{N})^2 + (\frac{v}{M})^2]$
- ▶ An approximation to the Fourier transform of the **continuous Laplacian**: $\nabla^2 = \frac{\partial^2}{\partial x^2} + \frac{\partial^2}{\partial y^2}$

Linear Image Filtering – Linear Image Denoising

- ▶ **Linear image denoising** is a process of smoothing noise **without** destroying image information
- ▶ Noise modeled as **additive** or **multiplicative**
- ▶ Consider **additive** noise now
- ▶ **Multiplicative** noise better handled by **non-linear** filtering

Linear Image Filtering – Additive White Noise Model

- ▶ Model **additive white noise** as an image N with random elements
- ▶ Can be **thermal noise**, **channel noise**, **sensor noise** etc.
- ▶ Noise may affect the **continuous image** before sampling:
$$J_C(x, y) = I_C(x, y) + N(x, y)$$

Linear Image Filtering – Additive White Noise Model

- ▶ Noise is **zero-mean** if limit of the average of P realizations of the noise image vanishes as $P \rightarrow \infty$:

$$\frac{1}{P} \sum_{p=1}^P N_{C,p}(x, y) \rightarrow 0 \text{ for all } (x, y) \text{ as } P \rightarrow \infty$$

- ▶ **On average**, noise falls close to 0

Linear Image Filtering – Spectrum of White Noise

- ▶ The **noise energy spectrum** is $N_C(\Omega, \Lambda) = \mathfrak{F}\{N_c(x, y)\}$
- ▶ If **noise is white**, then, on average, the **energy spectrum will be flat** (or 'white'):

$$\frac{1}{P} \sum_{p=1}^P |N_{C,p}(\Omega, \Lambda)| \rightarrow \eta \text{ for all } (\Omega, \Lambda) \text{ as } P \rightarrow \infty$$

- ▶ η^2 is called **noise power**

Linear Image Filtering – Linear Filtering

- ▶ **Objective: Remove** as much of the high frequency noise as possible while **preserving** as much image spectrum as possible
- ▶ Achieved by a LPF with large bandwidth (images are fairly wideband)

Linear Image Filtering – Digital White Noise

- ▶ Similar model for **digital** zero-mean additive white noise:

$$\underset{\text{observed}}{J} = \underset{\text{original}}{I} + \underset{\text{noise}}{N}$$

- ▶ **On average**, elements of N will be zero
- ▶ DFT of noisy image is: $\underset{\text{observed}}{\tilde{J}} = \underset{\text{original}}{\tilde{I}} + \underset{\text{noise}}{\tilde{N}}$
- ▶ **On average**, the noise DFT will contain a **broad band** of frequencies

Linear Image Filtering – Denoising with Average Filter

- ▶ To smooth an image: replace each pixel in a noisy image by the **average** of its $M \times M$ neighbors
- ▶ **Rationale:** Averaging elements **reduces the noise mean towards zero**
- ▶ How to pick window size M ? A **tradeoff** between noise smoothing and image smoothing
- ▶ Example: for a 512×512 image, **typical** window sizes range from 3×3 , 5×5 , ..., 15×15 (lots of smoothing)
- ▶ **Linear filtering** the image with zero padding affects the image and noise spectra in the same way:

$$K = H * J = H * I + H * N \implies \tilde{K} = \tilde{H} \otimes \tilde{I} + \tilde{H} \otimes \tilde{N}$$

Linear Image Filtering – Denoising with Ideal Low-Pass Filter

- Design an **ideal low-pass filter** in the DFT domain:

$$\tilde{H}(u, v) = \begin{cases} 1, & \text{if } \sqrt{u^2 + v^2} \leq U_{cutoff} \\ 0, & \text{otherwise} \end{cases}$$

- Possibly useful if **radial** frequency U_{cutoff} can be estimated in the original image

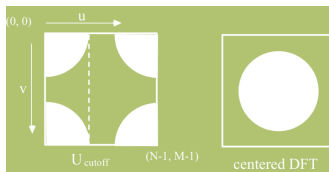


Figure: Ideal low-pass filter

Linear Image Filtering – Denoising with Gaussian Filter

- ▶ The **isotropic** Gaussian filter is an **effective** smoother:
$$\tilde{H}(u, v) = \exp[-2\pi^2\sigma^2(\frac{u^2+v^2}{N^2})]$$
- ▶ More weight given to “**closer**” neighbors
- ▶ DFT design: set **half-peak** bandwidth to U_{cutoff} and solve for σ :
$$\exp[-2\pi^2\sigma^2(\frac{U_{cutoff}^2}{N^2})] = \frac{1}{2} \implies \sigma = \frac{N}{\pi U_{cutoff}} \sqrt{\log\sqrt{2}} \approx 0.19(\frac{N}{U_{cutoff}})$$

Linear Image Filtering – Summary of Denoising Filters

- ▶ **Average filter:** Noise leakage through frequency ripple (spatial discontinuity)
- ▶ **Ideal low-pass filter:** Ringing from spatial ripple (frequency discontinuity)
- ▶ **Gaussian filter:** No ringing, no discontinuities and no leakage

Linear Image Filtering – Minimum Uncertainty

- ▶ Amongst **all real functions** and in any dimension, the no-ripple Gaussian functions **uniquely** minimize the uncertainty principle: $\left(\frac{\int |xf(x)|^2 dx}{\int |f(x)|^2 dx}\right)\left(\frac{\int |uf(u)|^2 du}{\int |f(u)|^2 du}\right) \geq \frac{1}{4}$
- ▶ Similar for y, v
- ▶ **Minimal simultaneous space-time duration**

Linear Image Filtering – Linear Image Deblurring

- ▶ Often a digital image is corrupted by a linear process
- ▶ **Motion blur, defocusing etc.** causes of blurring
- ▶ Blurring can be modeled using **linear convolution**:

$$J_C(x, y) = G_C(x, y) * I_C(x, y)$$
$$\implies \tilde{J}_C(\Omega, \Lambda) = \tilde{G}_C(\Omega, \Lambda) \tilde{I}_C(\Omega, \Lambda)$$

Linear Image Filtering – Digital Blur Function

- ▶ The sample image will be of the form: $J = G * I$ and $\tilde{J} = \tilde{G} \otimes \tilde{I}$
- ▶ The distortion G is **almost always low-pass**
- ▶ Goal: Use digital filtering to reduce blur - a **very** hard problem!

Linear Image Filtering – Inverse Filter Deblurring

- ▶ Often possible to **estimate** G
- ▶ **Physics** of the situation helps:
 - ▶ **Motion blur** along **one direction**. Determining this direction can help with filter design
 - ▶ Modulation transfer function (MTF) of camera system can be determined and a suitable **digital deblurring** filter can be designed

Linear Image Filtering – Deconvolution

- ▶ Reversing the linear blur G is **deconvolution**. It is the **inverse filter** of the distortion: $\tilde{G}_{inverse}(u, v) = \frac{1}{\tilde{G}(u, v)}$ if $\tilde{G}(u, v) \neq 0$ for any (u, v)
- ▶ The restored image is then: $\tilde{K} = \tilde{G}_{inverse} \otimes \tilde{G} \otimes \tilde{I} = \tilde{I}!$

Linear Image Filtering – Blur Estimation

- ▶ An **estimate** of blur G might be obtainable
- ▶ The inverse of **low-pass** blur is **high-pass**
- ▶ Designer must be **careful** at high frequencies

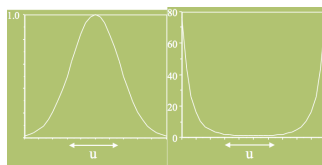


Figure: Inverting blurring filter

Linear Image Filtering – Blur Estimation

- ▶ Real world far from “ideal”
- ▶ Blur frequency response become **zero**
- ▶ Directly inverting $\tilde{G}(u, v)$ **meaningless** when $\tilde{G}(u, v) = 0$
- ▶ Frequencies **zeroed** by linear distortion are **irrecoverable**
- ▶ The best one can do is to reverse distortion at the **non-zero** values

Linear Image Filtering – Pseudo-Inverse Filter

- ▶ The **pseudo-inverse** filter is defined as:

$$\tilde{G}_{p-inverse}(u, v) = \begin{cases} 1/\tilde{G}(u, v), & \text{if } \tilde{G}(u, v) \neq 0 \\ 0, & \text{otherwise} \end{cases}$$

- ▶ No attempt to recover lost frequencies
- ▶ Psuedo-inverse set to 0 in regions of missing frequencies
- ▶ Spurious noise frequencies **eradicated**

Linear Image Filtering – Blur and Additive Noise

- ▶ Things get worse when I is distorted by linear blur G and additive noise N : $J = G * I + N$
- ▶ An example is image sent over a noisy communication channel
- ▶ The DFT: $\tilde{J} = \tilde{G} \otimes \tilde{I} + \tilde{N}$

Linear Image Filtering – Failure of Inverse Filters

- ▶ Filtering with a linear filter H will produce:

$$K = H * J = H * G * I + H * N \text{ or}$$

$$\tilde{K} = \tilde{H} \otimes \tilde{G} \otimes \tilde{I} + \tilde{H} \otimes \tilde{N}$$

- ▶ We have a problem:
 - ▶ Low-pass filter smooths over noise but doesn't remove blur
 - ▶ High-pass filter amplifies noise

Linear Image Filtering – Wiener Filter

- ▶ **Wiener filter** (after Norbert Wiener) is a minimum mean squared error (MMSE) filter
- ▶ For **blur** G and **white noise** N , $\tilde{G}_{Wiener}(u, v) = \frac{\tilde{G}(u, v)}{|\tilde{G}(u, v)|^2 + \eta^2}$
- ▶ Noise factor η usually not known - commonly estimated from flat image regions

Linear Image Filtering – Wiener Filter Rationale

- ▶ In case of no noise ($\eta = 0$), the filter reduces to
$$\tilde{G}_{Wiener}(u, v) = \frac{\tilde{G}(u, v)}{|\tilde{G}(u, v)|^2} = \frac{1}{\tilde{G}(u, v)}$$
- ▶ This is the highly desirable **inverse filter**
- ▶ In case of no blur, the filter reduces to $\tilde{G}_{Wiener}(u, v) = \frac{1}{1+\eta^2}$
- ▶ This essentially scales the variance to minimize MSE

Linear Image Filtering – Pseudo-Wiener Filter

- ▶ To handle zeroed frequencies, define a **pseudo-Wiener** filter:

$$\tilde{G}_{Wiener}(u, v) = \begin{cases} \frac{\tilde{G}(u, v)}{|\tilde{G}(u, v)|^2 + \eta^2}, & \text{if } \tilde{G}(u, v) \neq 0 \\ 0, & \text{otherwise} \end{cases}$$

- ▶ Noise in the zeroed frequency region **eradicated**