

- We learnt the equivalence between convolution and linear filtering.
- We reviewed two dimensional Fourier transforms of 2-d sequences.
- We discussed various properties of Fourier transforms and in particular we saw that the Fourier transform "converts" convolution to multiplication.
- Using the Fourier transform properties of Kronecker and Dirac delta functions we learnt about sampling and aliasing.

Definition of the 2-D Fourier Transform

The 2-D Fourier Transform of a 2-D sequence A, $\mathcal{F}(A)$ is defined as:

$$\mathcal{F}(\mathbf{A}) = F_A(w_1, w_2) = \sum_{m=-\infty}^{+\infty} \sum_{n=-\infty}^{+\infty} A(m, n) e^{-j(mw_1 + nw_2)} - \pi \le w_1, w_2 < \pi$$
(1)

The inverse 2-D Fourier Transform $\mathcal{F}^{-1}(\mathbf{A})$ is:

$$A(m,n) = \mathcal{F}^{-1}(\mathbf{A}) = \frac{1}{4\pi^2} \int_{-\pi}^{\pi} \int_{-\pi}^{\pi} F_A(w_1, w_2) e^{+j(mw_1 + nw_2)} dw_1 dw_2$$
(2)

 $\mathbf{2}$

 $A(m,n) \stackrel{\mathcal{F}}{\leftrightarrow} F_A(w_1,w_2)$

- A(m, n): two dimensional discrete sequence, m, n vary over integers
- $F_A(w_1, w_2)$: two dimensional 2π periodic function, w_1, w_2 vary in a continuum.

Fourier Transform and Convolution

•
$$\mathbf{C} = \mathbf{A} \otimes \mathbf{B}$$

$$C(m,n) = \sum_{k=-\infty}^{+\infty} \sum_{k=-\infty}^{+\infty} A(k,l)B(m-k,n-l)$$

$$F_C(w_1,w_2) = F_A(w_1,w_2)F_B(w_1,w_2)$$
(3)



Sampling and Aliasing

• The sampled sequence:

$$C(m,n) = A(S_1m, S_2n) \tag{4}$$

where $S_1, S_2 > 0$ are integers, has Fourier transform:

$$F_C(w_1, w_2) = \frac{1}{S_1 S_2} \sum_{k \in K(w_1)} \sum_{l \in L(w_2)} F_A(\frac{w_1}{S_1} - \frac{k2\pi}{S_1}, \frac{w_2}{S_2} - \frac{l2\pi}{S_2})$$
(5)

• For no aliasing to occur after sampling $F_A(w_1, w_2)$ must be:

$$F_A(w_1, w_2) = 0, \ \frac{\pi}{S_1} < |w_1| < \pi, \ \frac{\pi}{S_2} < |w_2| < \pi$$
 (6)

The Need for a "Computable" Fourier Transform

- The 2-D Fourier transform has many useful properties for the analysis of 2-D sequences and convolution.
- Unfortunately this Fourier transform can be computed explicitly for only some simple sequences.
 - The Fourier transform variables w_1, w_2 vary in a continuum. Thus the Fourier transform $F_A(w_1, w_2)$ of a sequence A(m, n) cannot be directly computed by a digital computer.
- We will now define "another" Fourier transform which is computable *and* enjoys similar nice properties.

The 2-D DFT for Finite Extent Sequences

Let A(m,n) be a finite extent sequence, i.e.,

$$A(m,n) \quad \begin{cases} \neq 0, & 0 \le m \le M_1 - 1, & 0 \le n \le N_1 - 1 \\ = 0 & \text{otherwise} \end{cases}$$

The $[M_1, N_1]$ point 2-D Discrete Fourier Transform (DFT) of A is defined as:

$$DF_A(k,l) = \sum_{m=0}^{M_1-1} \sum_{n=0}^{N_1-1} A(m,n) e^{-j(\frac{2\pi k}{M_1}m + \frac{2\pi l}{N_1}n)}, \quad k = 0, \dots, M_1 - 1, \ l = 0, \dots, N_1 - 1 \quad (7)$$

A(m,n) can be obtained "back" from $DF_A(k,l)$ via:

$$A(m,n) = \frac{1}{M_1 N_1} \sum_{k=0}^{M_1 - 1} \sum_{l=0}^{N_1 - 1} DF_A(k,l) e^{j(\frac{2\pi m}{M_1}k + \frac{2\pi n}{N_1}l)}$$
(8)



Remembering that A(m,n) is a finite extent sequence let us compare the definitions of the two Fourier transforms:

$$DF_A(k,l) = \sum_{m=0}^{M_1-1} \sum_{n=0}^{N_1-1} A(m,n) e^{-j(\frac{2\pi k}{M_1}m+\frac{2\pi l}{N_1}n)}, \quad k = 0, \dots, M_1 - 1, \ l = 0, \dots, N_1 - 1$$

$$F_A(w_1, w_2) = \sum_{m=0}^{M_1} \sum_{n=0}^{N_1} A(m,n) e^{-j(mw_1 + nw_2)} \qquad -\pi \le w_1, w_2 < \pi$$

• The 2-D DFT is a sampled version of $F_A(w_1, w_2)$, i.e.,

$$DF_A(k,l) = F_A(\frac{2\pi k}{M_1}, \frac{2\pi l}{N_1})$$
(9)

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Noting that $F_A(w_1, w_2)$ is periodic with 2π :

$$k = 0 \rightarrow w_1 = 0$$

$$k = M_1 - 1 \rightarrow w_1 = \frac{2\pi(M_1 - 1)}{M_1} = 2\pi - \frac{2\pi}{M_1} = -\frac{2\pi}{M_2}$$

$$k = M_1/2 \rightarrow w_1 = \pi \text{ if } M_1 \text{ even}$$

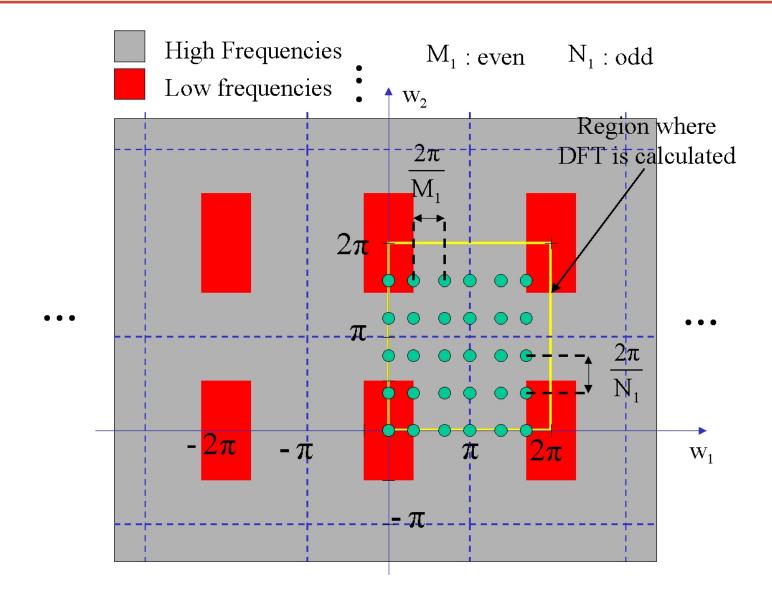
$$k = (M_1 - 1)/2 \rightarrow w_1 = \pi - \frac{\pi}{M_1} \text{ if } M_1 \text{ odd}$$

and similarly for l and w_2 .

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Example



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The 2-D DFT and the 2-D FT contd.

Now let us take a look at the inverse transforms:

$$A(m,n) = \frac{1}{M_1 N_1} \sum_{k=0}^{M_1-1} \sum_{l=0}^{N_1-1} DF_A(k,l) e^{j(\frac{2\pi m}{M_1}k + \frac{2\pi n}{N_1}l)} A(m,n) = \frac{1}{4\pi^2} \int_{-\pi}^{\pi} \int_{-\pi}^{\pi} F_A(w_1,w_2) e^{+j(mw_1+nw_2)} dw_1 dw_2$$

Note that the Fourier transform and its inverse are defined for *any* sequence whereas the DFT is *only* defined for finite extent sequences.

Using the inverse DFT and noting that $e^{j\frac{2\pi M_1}{M_1}} = 1$ we can see that:

$$A(m,n) = \frac{1}{M_1 N_1} \sum_{k=0}^{M_1-1} \sum_{l=0}^{N_1-1} DF_A(k,l) e^{j(\frac{2\pi m}{M_1}k + \frac{2\pi n}{N_1}l)} \times e^{j\frac{2\pi M_1}{M_1}} e^{j\frac{2\pi N_1}{N_1}} = \frac{1}{M_1 N_1} \sum_{k=0}^{M_1-1} \sum_{l=0}^{N_1-1} DF_A(k,l) e^{j(\frac{2\pi [m+M_1]}{M_1}k + \frac{2\pi [n+N_1]}{N_1}l)} = A(m+M_1, n+N_1) !$$
(10)

i.e., if we "forget" that A is finite extent, then the inverse DFT will reconstruct a *periodic* sequence which is called the periodic extension of A.



The 2-D DFT and Periodic Extensions

- The periodic extension property is normally not a problem since we can always do an inverse DFT for $0 \le m < M_1$, $0 \le n < N_1$.
- \bullet Consider however the DFT and its inverse for $\mathbf{C}=\mathbf{A}\otimes\mathbf{B}.$
 - We already know that if A and B are finite extent (A $(M_1 \times N_1)$, B $(M_2 \times N_2)$) then C is finite extent $(M_1 + M_2 1 \times N_1 + N_2 1)$.
 - $-DF_C(k,l)$ must therefore be computed via a $[M_1+M_2-1, N_1+N_2-1]$ point DFT.



The 2-D DFT and Convolution

Let A and B be finite extent sequences $(A (M_1 \times N_1), B (M_2 \times N_2))$.

• The $[M_1 + M_2 - 1, N_1 + N_2 - 1]$ point DFT of $C = A \otimes B$ is given as:

$$DF_C(k,l) = DF_A(k,l)_{[M_1+M_2-1,N_1+N_2-1]} \times DF_B(k,l)_{[M_1+M_2-1,N_1+N_2-1]}$$
(11)

- (If A and B are matrices, the $[M_1 + M_2 1, N_1 + N_2 1]$ point DFT of A can be computed by extending or "padding" A with zeros and similarly for B).
- Will suppress the $DF_X(k, l)_{[M_1+M_2-1, N_1+N_2-1]}$ notation with the understanding that the various DFTs are computed to the required point.
- Artifacts caused by *not* computing the DFTs to the required point are due to time aliasing.



• The DFT can be computed by a fast algorithm known as the FFT (Fast Fourier Transform). In matlab:

$$>> DFA = \mathsf{fft2}(A, M1, N1);$$

where M1, N1 denote the point to which DFT is computed.

• The convolution $\mathbf{C} = \mathbf{A} \otimes \mathbf{B}$ of finite extent sequences $\mathbf{A} (M_1 \times N_1)$ and $\mathbf{B} (M_2 \times N_2)$ can be computed *using* the fft algorithm via:

$$>> \quad DFA = {\rm fft} 2(A, M1 + M2 - 1, N1 + N2 - 1);$$

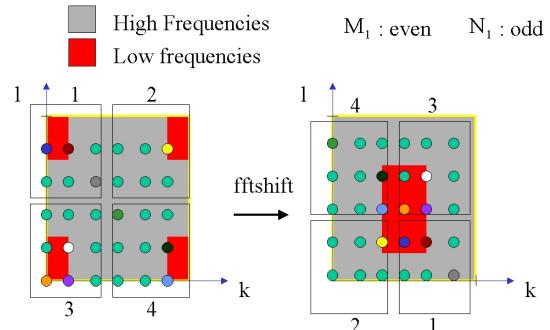
- >> $DFB = \text{fft}_2(B, M1 + M2 1, N1 + N2 1);$
- >> DFC = DFA. * DFB;

>>
$$C = ifft2(DFC, M1 + M2 - 1, N1 + N2 - 1);$$

where ifft2 denotes the inverse fft algorithm and $M1 = M_1$, $N1 = N_1$, etc.



DFT and fftshift



- The $[M_1, N_1]$ DFT of an $M_1 \times N_1$ image will have the low frequencies around $k = 0, k = N_1 1$ and $l = 0, l = M_1 1$ (see also earlier plot).
- When we plot image DFTs as images it is convenient to have the low frequencies at the center of the plot.
- This shift for viewing convenience can be done via the <code>ftshift</code> command in matlab.

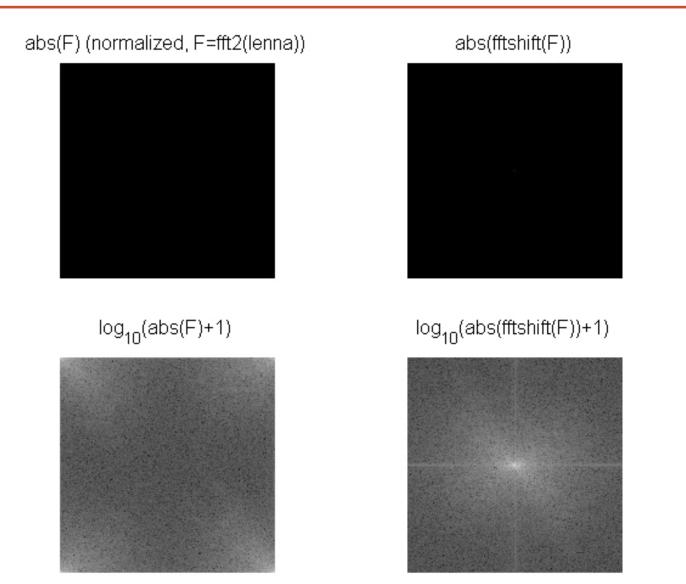


DFT and fftshift contd.

- >> DFA = fft2(A, M1, N1);
- >> DFA2 = fftshift(DFA);
- >> image(mynormalize(abs(DFA2)));
- fftshift in matlab will center the low frequencies for viewing convenience.
- Note that $DFA2 \neq DFA$ and thus ifft2 $(DFA2, M1, N1) \neq ifft2(DFA, M1, N1)$.
- *DFA* can be obtained from *DFA*² by doing *another* fftshift, i.e., *DFA* = fftshift(*DFA*²).

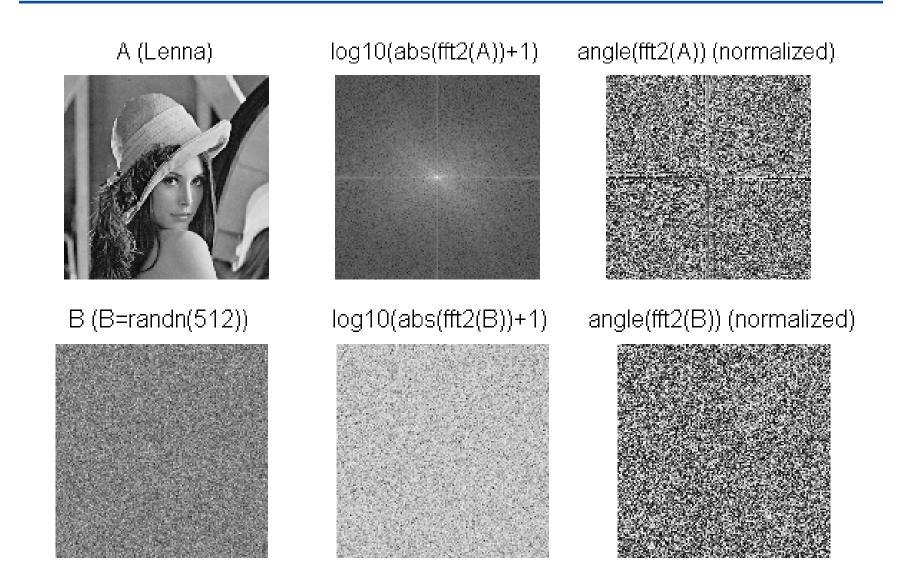


Example





DFTs of Natural Images



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The importance of low frequency coefficients of image DFTs can be demonstrated as follows:

• Let A be $M_1 \times N_1$. Let

$$\mathcal{R}_1 = \{i | (0 \le i \le W_1) \text{ or } (M_1 - W_1 \le i \le M_1 - 1)\}$$

$$\mathcal{R}_2 = \{i | (0 \le i \le W_2) \text{ or } (N_1 - W_2 \le i \le N_1 - 1)\}.$$

Define a $(2W_1 + 1) \times (2W_2 + 1)$ window "around" the low frequencies by

$$w(k,l) = \begin{cases} 1 & k \in \mathcal{R}_1 \text{ and } l \in \mathcal{R}_2 \\ 0 & \text{otherwise} \end{cases}$$
(12)

- Consider $DF_C(k, l) = DF_A(k, l)w(k, l)$. This "keeps" $(2W_1 + 1) \times (2W_2 + 1)$ DFT coefficients and zeros out the rest.
- We are interested in the mean squared error given by $1/(M_1N_1) \sum_{m=0}^{M_1-1} \sum_{n=0}^{N_1-1} (A(m,n) C(m,n))^2$.
- Note that as W_1, W_2 increase we are adding more and more high coefficients to the set of coefficients that we keep.



Low Frequency Window



w for $W_1 = W_2 = 100$ (normalized and fftshifted)

The window can be implemented in matlab via

>>
$$r1 = \operatorname{zeros}(M1, 1);$$

>> $r1(1:W1+1, M1 - W1 + 1:M1) = 1;$
>> $r2 = \operatorname{zeros}(N2, 1);$
>> $r2(1:W2+1, N1 - W2 + 1:N1) = 1;$
>> $w = r1 * r2';$

Given $DF_C(k,l) = DF_A(k,l)w(k,l)$, matlab may make small numerical errors in the inverse transform leading to complex valued images. C(m,n) is best reconstructed via:

$$>> C = \operatorname{real}(\operatorname{ifft2}(w.*DFA));$$

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Example

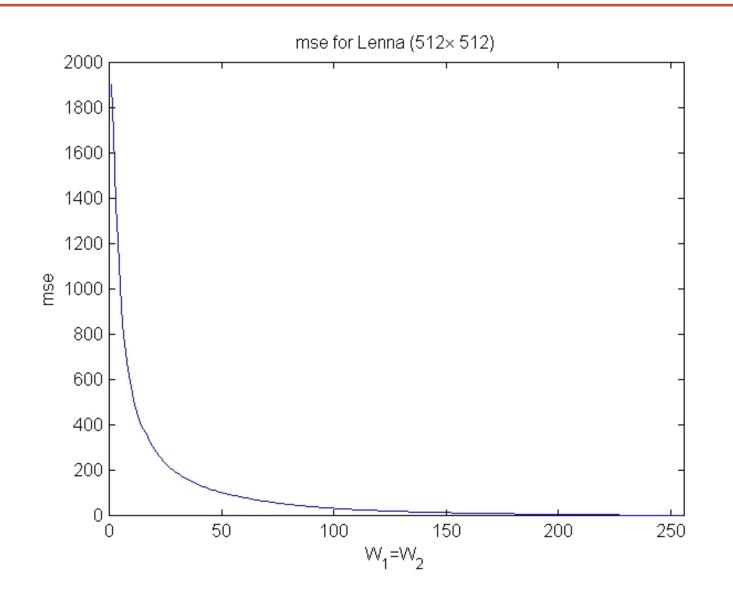


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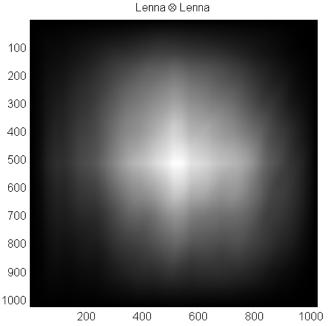
Example



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- Since the DFT can be computed with a fast algorithm it may be beneficial to do the convolution of two sequences A $(M_1 \times N_1)$ and B $(M_2 \times N_2)$ via $[M_1 + M_2 + 1, N_1 + N_2 + 1]$ point DFTs.
- However, speed improvements are only possible if both sequences have large dimensions. Otherwise convolutions are better implemented via the convolution sum.





- In this lecture we learnt the 2-D DFT of two dimensional finite extent sequences.
- We learnt how to calculate convolutions using DFTs.
- We learnt about basic properties of the DFTs of natural images.

Homework VII

- 1. Calculate the DFT of your image. Show the magnitude and phase both before and after using fftshift. Use the log10 point function on magnitude plots and normalize as necessary.
- 2. Subsample your image by 2 in each direction and calculate the DFT of the result in two ways:
 - (a) Since the subsampled image has half the dimensions of the original, calculate the DFT to the *point* of the reduced dimensions.
 - (b) Calculate the DFT to the point of the original dimensions.

Show magnitude and phase plots for both. Compare the results to 1. above. Can you explain the differences?

- 3. Calculate the convolution of your image with itself by using DFTs.
- 4. Do the processing I did on Pages 19-20. Show the resulting images and the mse plot.

References

[1] A. K. Jain, Fundamentals of Digital Image Processing. Englewood Cliffs, NJ: Prentice Hall, 1989.