

Homework #4, EECS 556, W21. Due **Thu. Feb. 18**, by 9:00AM

Skills and Concepts

- 2D discrete-space (DS) signals and systems
- DSFT
- 2D filter design

Problems

1. [15] The response of a 2D DS system \mathcal{S} to the unit step input $\text{step}_2[m, n]$ is known to be $s[m, n]$. For each of the following cases, determine the class of input images $f[m, n]$ for which we can determine the output $g[m, n]$ in terms of $s[m, n]$. For each input in the class, express the output in terms of $s[m, n]$.
- (a) [5] \mathcal{S} is **linear**, but not **shift invariant** (SI)
- (b) [5] \mathcal{S} is SI, but not linear
- (c) [5] \mathcal{S} is LSI
- Hint: express $\delta_2[m, n]$ in terms of $\text{step}_2[m, n]$.

2. [10] The course notes give a somewhat messy expression for the **sum-pooling** layer used in some **CNN** models. Find a concise expression using a combination of down-sampling notation and convolution with an appropriate filter. Is your filter separable? Hint. The filter is quite small.

3. [15]
- (a) [10] Find a (2D DS) FT property for **up-sampling** (using insertion of zeros) by a factor of 2, *i.e.*, if $g[m, n] = f_{\uparrow 2}[m, n]$ then relate $G(\Omega_1, \Omega_2)$ and $F(\Omega_1, \Omega_2)$.
- (b) [5] Suppose the image with spectrum $F(\Omega_1, \Omega_2) = \text{rect}\left(\frac{\sqrt{p(\Omega_1)^2 + p(\Omega_2)^2}}{2\pi}\right)$ is up-sampled by a factor of 2, where $p(\Omega) = (\Omega + \pi) \bmod 2\pi - \pi$ ensures that $F(\Omega_1, \Omega_2)$ is appropriately periodic. Carefully sketch the resulting spectrum after up-sampling.
- (c) [0] Do you see evidence of aliasing effects? If so, how would you remove them?

4. [10] In continuous space, if $g(x, y) = f(x, y) ** h(x, y) \xleftrightarrow{\mathcal{F}_2} G(\nu_x, \nu_y) = F(\nu_x, \nu_y) H(\nu_x, \nu_y)$, then we have the following “combined scaling/convolution property” of the 2D FT:

$$f(2x, 2y) ** h(2x, 2y) \xleftrightarrow{\mathcal{F}_2} \frac{1}{4} F\left(\frac{\nu_x}{2}, \frac{\nu_y}{2}\right) \frac{1}{4} H\left(\frac{\nu_x}{2}, \frac{\nu_y}{2}\right) = \frac{1}{16} G\left(\frac{\nu_x}{2}, \frac{\nu_y}{2}\right).$$

What about in discrete space? If $g[m, n] = f[m, n] ** h[m, n] \xleftrightarrow{\text{DSFT}} G(\Omega_1, \Omega_2) = F(\Omega_1, \Omega_2) H(\Omega_1, \Omega_2)$, then is there a simple relationship between $G(\Omega_1, \Omega_2)$ and the spectrum of $y[m, n] \triangleq f_{\downarrow 2}[m, n] ** h_{\downarrow 2}[m, n] = f[2m, 2n] ** h[2m, 2n]$? If so, find it. If not, at least find an expression for the spectrum of $y[m, n]$ in terms of $F(\Omega_1, \Omega_2)$ and $H(\Omega_1, \Omega_2)$.
Hint: first derive a **down-sampling** relation for the 2D DSFT, for which the expression $[1 + (-1)^n]/2$ will be useful.
Hint: arguments like $\frac{\Omega_1}{2} - \pi$ will appear as part of your answer.

5. [10] Using the **Hankel transform** pair

$$\frac{\sin(2\pi r)}{r} \xleftrightarrow{\mathcal{F}_2} \frac{\text{rect}\left(\frac{\rho}{2}\right)}{\sqrt{1 - \rho^2}},$$

determine analytically the **impulse response** $h[m, n]$ of the digital filter having **frequency response**

$$H(\Omega_1, \Omega_2) = \begin{cases} \frac{1}{\sqrt{1 - 4\tilde{\Omega}_1^2 - 4\tilde{\Omega}_2^2}}, & \tilde{\Omega}_1^2 + \tilde{\Omega}_2^2 \leq 1/4 \\ 0, & \text{otherwise,} \end{cases}$$

where $\tilde{\Omega} = (\Omega + \pi) \bmod 2\pi - \pi$.

Optional questions:

- Is $h[m, n]$ weak-sense circularly symmetric?
- Is $h[m, n]$ strong-sense circularly symmetric?
- Is $H(\Omega_1, \Omega_2)$ circularly symmetric?

-
6. [10] Because there are many good 1D FIR filter design methods, this problem explores how one might build 2D FIR filters from 1D designs.
- (a) [10] Given a 1D symmetric filter $h[n]$ with frequency response $H(\Omega)$, one way to specify a 2D frequency response is by $H(\Omega_1, \Omega_2) = H(\Omega) \Big|_{\cos \Omega = \frac{1}{2}(-1 + \cos \Omega_1 + \cos \Omega_2 + \cos \Omega_1 \cos \Omega_2)}$. For the specific case $h[n] = 2\delta[n] + \delta[n-1] + \delta[n+1]$, determine the resulting 2D impulse response $h[m, n]$.
- (b) [0] Is the resulting filter separable?
- (c) [0] An alternative approach would be to use: $H(\Omega_1, \Omega_2) = H\left(\sqrt{\Omega_1^2 + \Omega_2^2}\right)$ with appropriate modulo 2π consideration. What would be the primary advantage and primary disadvantage of this alternative?
-
7. [80] The purpose of this problem is to explore the design of an (approximately) rotationally invariant filter, *i.e.*, a filter whose frequency response is approximately circularly symmetric.
- (a) [10] Consider the 1D filter with impulse response $h[n] = [-1/4 \ 0 \ 1/2 \ 0 \ -1/4]$. Determine and plot the frequency response of this 1D filter, and describe qualitatively what kind of filter it is.
- (b) [10] Specify the (circularly-symmetric) frequency response $H(\Omega_1, \Omega_2)$ of a 2D filter with equivalent behavior.
- (c) [10] Display that frequency response using `imagesc` or `jim`. (As usual, make sure DC is at the center, and remember to include a `colorbar`.)
- (d) [10] Find the impulse response $h[m, n]$ of that filter. (You may try to do this analytically, or just do it numerically following the example in the notes.) Clearly record the central 5×5 part of $h[m, n]$ for grading.
- (e) [10] Extract the central 5×5 portion of $h[m, n]$ (*i.e.*, truncate the impulse response) and compute the frequency response $H_t(\Omega_1, \Omega_2)$ of the truncated filter. Display this frequency response using `imagesc` or `jim`.
- (f) [10] Use the `contour` command (with an appropriate optional argument to show 8 contour lines at levels from 0.1 to 1) to overlay (use `hold` or `contour!`) the contours of the truncated frequency response $H_t(\Omega_1, \Omega_2)$ with those of the ideal response $H(\Omega_1, \Omega_2)$.
- (g) [20] Overlay plots of the central horizontal profiles $H(\Omega, 0)$ and $H_t(\Omega, 0)$ vs Ω .
- (h) [0] Do the contours and profiles agree? Explain why or why not. What do you conclude about FIR design of circularly symmetric filters? Experiment with other amounts of truncation, or with non-rectangular truncation.

Optional problems

8. [0] This problem is an elementary preview of the principles underlying transform coding with truncation. A typical digital image $x[m, n]$ has a spectrum that decays with increasing spatial frequency. As a concrete model, suppose that $|X(\Omega_1, \Omega_2)| = \begin{cases} A e^{-\alpha\sqrt{\Omega_1^2 + \Omega_2^2}}, & \Omega_1^2 + \Omega_2^2 \leq \pi^2 \\ 0, & \text{otherwise.} \end{cases}$
- Suppose that we truncate the tails of this spectrum by as follows: $Y(\Omega_1, \Omega_2) = \begin{cases} X(\Omega_1, \Omega_2), & \sqrt{\Omega_1^2 + \Omega_2^2} \leq \pi/10 \\ 0, & \text{otherwise,} \end{cases}$ and then reconstruct the signal $y[m, n]$ by an inverse 2D DSFT.
- (a) [0] For $\alpha = 5$, evaluate the normalized root mean-squared error (NRMSE) $\sqrt{\frac{\sum_{m=-\infty}^{\infty} \sum_{n=-\infty}^{\infty} |y[m, n] - x[m, n]|^2}{\sum_{m=-\infty}^{\infty} \sum_{n=-\infty}^{\infty} |x[m, n]|^2}}$.
- (b) [0] This procedure retains only the fraction $\pi(\pi/10)^2/(4\pi^2)$, or about 1%, of the frequency components. So it can be thought of as 100-fold data compression. Discuss the distortion caused by this drastic data reduction.
-

9. [0]
- (a) [0] If $h_l[n]$ is the impulse response of a 1D FIR lowpass filter, then a simple way to design a 1D FIR highpass filter $h[n]$ is by letting $h[n] = (-1)^n h_l[n]$. Show that $h[n]$ is a highpass filter.
- (b) [0] Suppose $h_l[m, n]$ is the impulse response of a good 2D FIR lowpass filter. A natural extension of the above method is to try designing a 2D FIR highpass filter by letting $h[m, n] = (-1)^m (-1)^n h_l[m, n]$. Is this a good method of designing a 2D highpass filter?
-

10. [0] (a) [0] Find the frequency response $H(\Omega_1, \Omega_2)$ and impulse response $h[m, n]$ of a FIR highpass filter whose frequency response satisfies the following:

$$H(\Omega_1, \Omega_2) = \begin{cases} 0, & \Omega_1 = \Omega_2 = 0 \\ 1, & (\Omega_1, \Omega_2) \in \{(-\pi, \pm\pi), (0, \pm\pi), (\pi, \pm\pi), (\pm\pi, 0)\} \\ ?, & \text{otherwise.} \end{cases}$$

Try to choose the “?” part of $H(\Omega_1, \Omega_2)$ so that $h[m, n]$ is as simple as possible. Hint: a 3×3 filter suffices.

- (b) [0] Suppose that processing an input image $x[m, n]$ with the above filter $h[m, n]$ yields the output image $y[m, n]$. Determine $\sum_{m=-\infty}^{\infty} \sum_{n=-\infty}^{\infty} y[m, n]$.
- (c) [0] Suppose $x[m, n]$ is real and nonnegative; from your preceding answer, describe how $y[m, n]$ will appear on a display where all negative values of $y[m, n]$ appear as black (like zero).