

Public Shift Theorem

$$F_{R} = \sum_{n=0}^{N-1} f_{n} \overline{W}^{nk}$$
 $k = 0, ..., N-1$

VERSION I

$$G_{k} = \sum_{n=0}^{N-1} g_{n} \overline{W}^{nk}$$

$$= \sum_{n=0}^{N-1} f_{n-s} \overline{W}^{nk} \times \overline{W}^{sk} \overline{W}^{-sk}$$

$$= \overline{W}^{sk} \sum_{n=0}^{N-1} f_{n-s} \overline{W}^{k(n-s)}$$

Change of variables: let
$$m = n-s$$

= $\overline{W} \cdot k \sum_{m=-s}^{N-1-s} f_m \overline{W}^{mk}$

$$= \overline{W}^{sk} \sum_{m=0}^{N-1} f_m \overline{W}^{mk} \quad \text{since } f \in W \text{ are } N\text{-periodic}$$

$$= \overline{W}^{sk} F_k \quad k=0,...,N-1$$

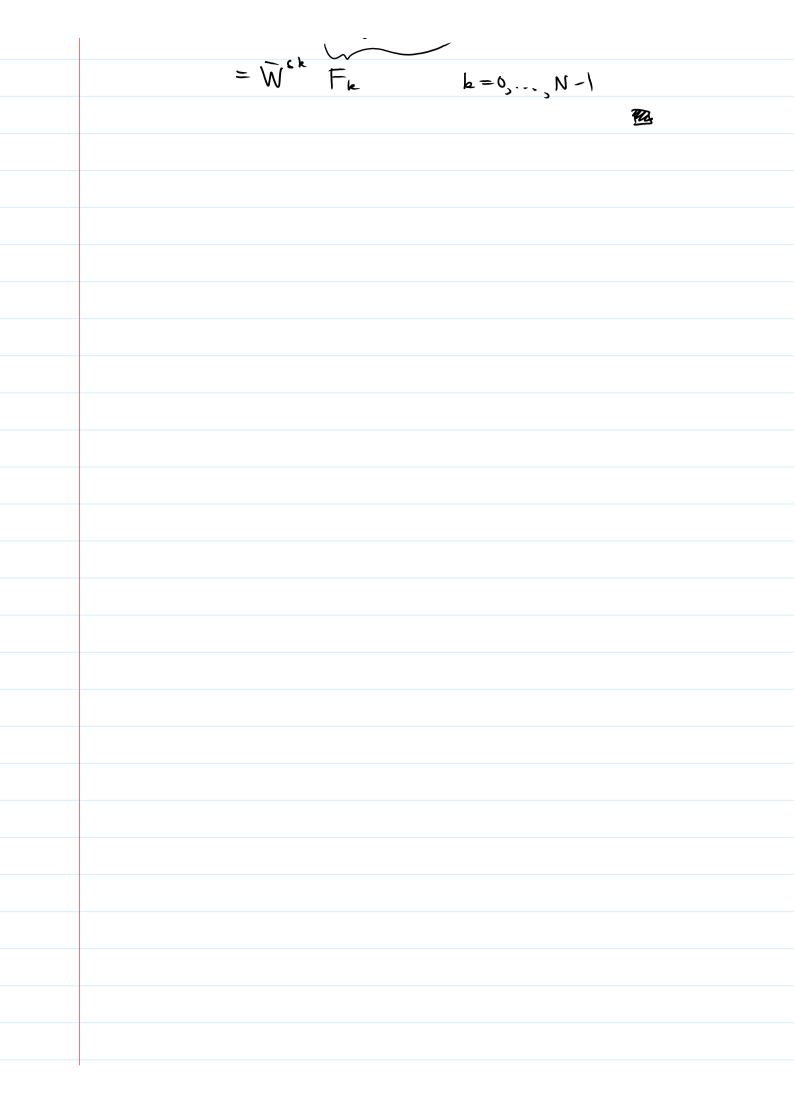
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VERSION I

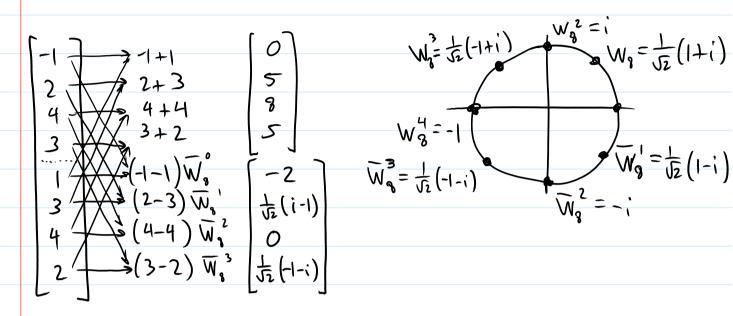
Change of vars: m=n-s => n=m+s

$$C_k = \sum_{m=-s}^{N-1-s} f_m \overline{W}^{(m+s)} k$$

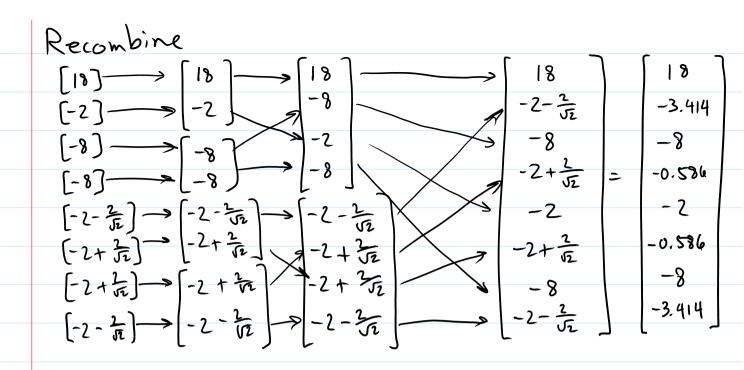
$$= \widetilde{W}^{ck} F_{L} \qquad b = 0 \qquad N(-1)$$



Public FFT v2 Solution



$$\begin{bmatrix} -2 \\ -\frac{1}{\sqrt{2}} \end{bmatrix} \qquad \begin{bmatrix} -2 - \frac{1}{\sqrt{2}} \\ -\frac{1}{\sqrt{2}} \end{bmatrix} \qquad \begin{bmatrix} -2 - \frac{1}{\sqrt{2}} \\ -2 + \frac{1}{\sqrt{2}} \end{bmatrix} \qquad \begin{bmatrix} -2 + \frac{1}{\sqrt{2}} \\ -2 + \frac{1}{\sqrt{2}} \end{bmatrix} \qquad \begin{bmatrix} -2 + \frac{1}{\sqrt{2}} \\ -2 - \frac{1}{\sqrt{2}} \end{bmatrix} \qquad \begin{bmatrix} -2 - \frac{1}{\sqrt{2}} \\ -2 - \frac{1}{\sqrt{2}} \end{bmatrix}$$



```
Public myDCT Solutions
 function Fdct = myDCT(f)
    % Perform an even extension of f
   fe = EvenExtension(f);
    % 2D-DFT yields purely real Fourier coefficients
    Fe = real(fft2(fe));
    % We don't need to keep them all because of symmetry
    % (recall that the input is real-valued).
    % We only need to keep the same array size as the input.
    Fdct = IEvenExtension(Fe);
 function f = myIDCT(Fdct)
    % Even extension of Fdct to restore symmetric Fourier coefficients
    Fe = EvenExtension(Fdct);
    % 2D-IDFT to get even-extended f
   f = ifft2(Fe);
    % Extract only one non-redundant part, the same size as the input.
    f = IEvenExtension(f);
```

```
Public JPEG Solutions
4(a) function G = myJPEGCompress(f, T, D)
         [h,w] = size(f); % returns the width and height of f
        % Process tiles and blocks, using loops
        wtiles = 1:T:(w-mod(w,T));
        htiles = 1:T:(h-mod(h,T));
        G = zeros(length(htiles)*D, length(wtiles)*D);
        C = 1;
        for c = wtiles
          R = 1;
          for r = htiles
             % Extract a TxT tile from the image.
             ftile = f(r:(r+T-1),c:(c+T-1));
             % Perform the DCT on the tile
             F = myDCT(ftile);
             % Extract a DxD block of coefficients
             Fblock = F(1:D,1:D);
             % Store the block of coefficients in the output G
             G(R:(R+D-1),C:(C+D-1)) = Fblock;
             R = R + D;
          end
          C = C + D;
        end
     function [f] = myJPEGDecompress(G, T, D)
4(b)
        n hblocks = size(G,1)/D;
```

```
n_{wblocks} = size(G,2)/D;
        f = zeros(T*n_hblocks, T*n_wblocks);
        wblocks = 1:T:size(f,2);
        hblocks = 1:T:size(f,1);
        % Process tiles and blocks, including loops
        C = 1:
        for c = wblocks
          R = 1;
          for r = hblocks
             % Embed DCT coefs in a TxT array
             Fsub = zeros(T,T);
             Fsub(1:D,1:D) = G(R:(R+D-1),C:(C+D-1));
             % Call IDCT
            fe = myIDCT(Fsub);
             % Copy tile into image
            f(r:(r+T-1),c:(c+T-1)) = fe;
             R = R + D;
          end
          C = C + D;
        end
4(c)
      % image compression.m
      f = imread('../house.jpg');
      f = double(f(:,:,1));
      T = 25;
      %% Display the original, and the uncompressed
```

```
figure(1);
D = 11; % compression ratio of 5:1
G = myJPEGCompress(f, T, D);
fc = myJPEGDecompress(G, T, D);
subplot(2,2,1); % top-left in a 2x2 grid of images
imshow(fc, [0 255]);
title(['Compression Ratio = 'num2str(round(numel(f)/numel(G))) ':1']);
D = 8; % compression ratio of 10:1
G = myJPEGCompress(f, T, D);
fc = myJPEGDecompress(G, T, D);
subplot(2,2,2); % top-right
imshow(fc, [0 255]);
title(['Compression Ratio = 'num2str(round(numel(f)/numel(G))) ':1']);
D = 5; % compression ratio of 25:1
G = myJPEGCompress(f, T, D);
fc = myJPEGDecompress(G, T, D);
subplot(2,2,3); % bottom-left
imshow(fc, [0 255]);
title(['Compression Ratio = 'num2str(round(numel(f)/numel(G))) ':1']);
D = 3; % compression ratio of 70:1
G = myJPEGCompress(f, T, D);
fc = myJPEGDecompress(G, T, D);
subplot(2,2,4); % bottom-right
imshow(fc, [0 255]);
title(['Compression Ratio = 'num2str(round(numel(f)/numel(G))) ':1']);
```

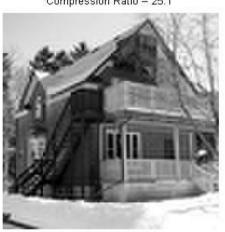
Compression Ratio = 5:1



Compression Ratio = 10:1



Compression Ratio = 25:1



Compression Ratio = 69:1



```
Public Filter Recording Solution
   %% filter audio.m
   %Load some audio
   blah = importdata('recording.wav');
   f = blah.data; % Rename the variables to our liking
   Omega = blah.fs; % Sampling rate (samples per second, or Hz)
   clear blah: % We don't need this anymore
   % Derive some useful variables
   N = length(f); % Number of samplesL = N / Omega; % Clip duration in seconds
   t = (0:(N-1))' * L/N; % Time axis labels
   omega = ifftshift((-N/2):(N/2-1))'/L;
   %omega = fftshift( (-(N-1)/2 - 1):((N-1)/2 - 1))' / L; % Frequency axis labels
   % Plot time domain signal f
   figure(1);
   plot(f); title('Sound');
   xlabel('Time (seconds)');
   ylabel('Amplitude');
   %% Listen to the sound
   % If you are running Matlab locally, you can play the sound using
   % sound(f, Omega);
   % You can also save a way file and download and listen to it.
   audiowrite('sound file.wav', real(f), Omega);
   %% === YOUR CODE HERE ===
   %% (a)
   % Take DFT of audio signal
```

```
F = fft(f); % DFT of f
% Plot modulus of frequency domain, abs(F)
figure(2);
plot(omega, abs(F));
title('Frequency domain F');
xlabel('Freq (Hz)');
                                         700
ylabel('Modulus');
print('a DFT of f.png', '-dpng');
                                        sninboy
%% (b)
                                         200
% Choose a frequency threshold
                                          0 <sup>∟</sup>
-2.5
                                                          Freq (Hz)
tau = 2250; % Threshold frequency
% Extract the high frequencies (above tau)
% Apply the threshold
mask = abs(omega)<tau;</pre>
F lowpass = F.* mask;
% Plot the filtered Fourier coefficients
figure(3);
plot(omega, abs(F_lowpass));
title('Low-Pass');
xlabel('Frequency (Hz)');
ylabel('Modulus');
print('b DFT of filtered f.png', '-dpng');
%% (c)
%% Reconstruct the low-pass and high-pass signals
f lowpass = ifft(F lowpass);
sound(10*f lowpass, Omega);
%audiowrite('sound file.wav', real(f), Omega);
                                      0.025
```

