Data Compression

# Why we need data compression?

- Fax machine: 40000 DPSI => 4 million dots per page
- 56 KBPS modem, time to transmit = ?
- Video: 30 pictures per second
- Each picture = 200,000 dots or pixels
- 8-bits to represent each primary color
- Bits required for one picture = ?
- Two hour movie requires = ?

# Introduction

- Compression is a way to reduce the number of bits in a frame but retaining its meaning.
- Decreases space, time to transmit, and cost
- Technique is to identify *redundancy* and to eliminate it
- If a file contains only capital letters, we may encode all the 26 alphabets using 5-bit numbers instead of 8-bit ASCII code

#### Introduction

 If the file had n-characters, then the savings = (8n-5n)/8n => 37.5%

- Not all the characters appear with same frequency, some are more prevalent than the others
- Frequently appearing characters could be assigned shorter codes than the others => results in reduced number of bits
- Such codes are examples of frequency dependent code

• Huffman code: (illustrated with a manageable example)

| Letter | Frequency (%) |  |  |
|--------|---------------|--|--|
| А      | 25            |  |  |
| В      | 15            |  |  |
| С      | 10            |  |  |
| D      | 20            |  |  |
| E      | 30            |  |  |

- Huffman code: Code formation
- Assign weights to each character
- Merge two lightest weights into one root node with sum of weights (if multiple? Not unique code)
- Repeat until one tree is left
- Traverse the tree from root to the leaf (for each node, assign 0 to the left, 1 to the right)

- Huffman code: Code Interpretation
- No prefix property: code for any character never appears as the prefix of another code (Verify)
- Receiver continues to receive bits until it finds a code and forms the character
- 01110001110110110111 (extract the string)

• Arithmetic compression: is based on Interpreting a character-string as a single real number

| Letter | Frequency (%) | Subinterval [p, q] |
|--------|---------------|--------------------|
| А      | 25            | [0, 0.25]          |
| В      | 15            | [0.25, 0.40]       |
| С      | 10            | [0.40, 0.50]       |
| D      | 20            | ?                  |
| E      | 30            | ?                  |

- Arithmetic compression: Coding 'CABAC'
- Generate subintervals of decreasing length, subintervals depend uniquely on the string's characters and their frequencies.
- Interval [x, y] has width w = y x, the new interval based on [p, q] is x = x + w.p, y = x + w.q
- Step 1: 'C' 0......0.4....0.5.....1
   based on p = 0.4, q = 0.5

• Step 2: 'A' 0.4.....0.425.....0.5 based on p = 0.0, q = 0.25

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Step 3: 'B'
0.4....0.40625....0.41....0.425
based on p = 0.25, q = 0.4
Step 4: 'A'
Step 5: 'C'
...0.406625... 0.4067187...
Final representation (midpoint)?
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#### • Arithmetic compression: Extracting 'CABAC'

| Ν      | Interval[p, q] | Width | Character | N-p    | (N-p)/width |
|--------|----------------|-------|-----------|--------|-------------|
| 0.4067 | 0.4 – 0.5      | 0.1   | С         | 0.0067 | 0.067       |
| 0.067  | 0 – 0.25       | 0.25  | А         | 0.067  | 0.268       |
| 0.268  | 0.25 – 0.4     | 0.15  | В         | 0.018  | 0.12        |
| ?      |                |       |           |        |             |
| 2      |                |       |           |        |             |

When to stop? A terminal character is added to the original character set and encoded. During decompression, once it is encountered the process stops.

- Huffman code requires:
  - frequency values
- bits are grouped into characters or units Many items do not fall into such category
  - machine code files
  - facsimile Data (bits corresponding to light or dark area of a page)
  - video signals

- For such files, RLE is used.
- 'Instead of sending long runs of '0's or '1's, it sends only how many are in the run.'
- 70%-80% space is white on a typed character space, so RLE is useful.

Α

- Runs of the same bit
- In facsimile Data, there are many '0's (white spots) -> transmit the run-length as fixed size binary integer
- Receiver generates proper number of bits in the run and inserts the *other* bit in between
- 14 zeros, 1, 9 zeros, 11, 20 zeros, 1, 30 zeros, 11, 11 zeros (number of zeros encoded in 4bits)

- Runs of the same bit
- Code: 1110 1001 0000 1111 0101 1111 1111
   0000 0000 1011
- (next value after 1111 is added to the run)
- SAVINGS IN BITS: ?
- If the stream started with '1' instead?
- Best when there are many long runs of zeros, with increased frequency of '1's, becomes less efficient.

- Runs with different characters
- Send the actual character with the run-length
- HHHHHHUFFFFFFFYYYYYYYDGGGGGG
- code = 7, H, 1, U, 9, F, 11, Y, 1, D, 5, G
- SAVINGS IN BITS (considering ASCII): ?

- Facsimile Compression
- ITU standard (A4 document, 210 by 297 mm)
- 1728 pixels per line
- If 1 bit for each pixel, then over 3 million bits for each page
- A typical page contains many consecutive white or black pixels -> RLE

- Run lengths may vary from 0 to 1728 -> many
   Possibilities and inefficiency with a fixed size code
- Some runs occur more frequently than others, e.g. most typed pages contain 80% white pixels, spacing between letters is fairly consistent
- => probabilities of certain runs are predictable
- => Frequency dependent code on run lengths

• Some Facsimile compression codes (Terminating, less than 64)

Pixels in the run Code: White Code: Black

()

 Some Facsimile compression codes (Make up, greater than or equal to 64)

Pixels in the run Code: White Code: E

641101112810010

Code: Black 0000001111 000011001000

- 256
- 512

# 129 white: Savings:

No-prefix property, better compression for long-runs

# **Relative Encoding**

- Relative Encoding:
- Some applications may not benefit from the above: video image -> little repetitive within, but much repetition from one image to the next
- Differential encoding is based on coding only the difference from one to the next

# **Relative Encoding**

• Relative Encoding:

| • 1234                           | 1 <mark>3</mark> 3 4  | 0100       |  |  |
|----------------------------------|-----------------------|------------|--|--|
| 2537                             | 2537                  | 0000       |  |  |
| 3648                             | 3647                  | 000-1      |  |  |
| 4759                             | <mark>3</mark> 759    | -1000      |  |  |
| 1 <sup>st</sup> Frame            | 2 <sup>nd</sup> Frame | Difference |  |  |
| Resulting difference can be RLE. |                       |            |  |  |

#### Image Representation

- BW pixels each represented by 8-bit level
- Color composed of R, G, B primaries, each is represented by 8-bit level
- -> Each color pixel can be represented by one of 2<sup>8</sup>.2<sup>8</sup>.2<sup>8</sup> = 2<sup>24</sup> colors

VGA screen: 640 \* 480 pixels

-> 640 \* 480 \* 24 = 7, 372, 800 bits

# **Image Compression**

- JPEG compression both for grayscale and color images
- Previous compression methods were lossless it was possible to recover all the information from the compressed code
- JPEG is lossy: image recovered may not be the same as the original

• It consists of three phases: Discrete Cosine Transform (DCT), Quantization, Encoding.

• DCT:

Image is divided into blocks of 8\*8 pixels For grey-scale images, pixel is represented by 8-bits

For color images, pixel is represented by 24bits or three 8-bit groups

- DCT takes an 8\*8 matrix and produces another 8\*8 matrix.
- T[i][j] = 0.25 C(i) C(j) ∑ ∑ P[x][y] Cos (2x+1)iπ/16 \* Cos (2y+1)jπ/16

$$i = 0, 1, ..., 7, j = 0, 1, ..., 7$$

C(i) =  $1/\sqrt{2}$ , i =0

= 1 otherwise

T contains values called 'Spatial frequencies'

- Spatial frequencies directly relate to how much the pixel values change as a function of their positions in the block
- T[0][0] is called the DC coefficient, related to average values in the array/matrix, Cos 0 = 1
- Other values of T are called AC coefficients, cosine functions of higher frequencies

- Case 1: all P's are same => image of single color with no variation at all, AC coefficients are all zeros.
- Case 2: little variation in P's => many, not all, AC coefficients are zeros.
- Case 3: large variation in P's => a few AC coefficients are zeros.

P-matrix

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• 20 30 40 50 60 ... 720 -182 0 -19 0 ... 30 40 50 60 70 40 50 60 70 80 0 0 0 0 0 50 60 70 80 90 -19 0 0 0 0 60 70 80 90 100 0 0 0 0 0

**T-matrix** 

- -182 0 0 0 0

'Uniform color change and little fine detail, easier to compress after DCT'

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P-matrix

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100 150 50 100 100 ...
835 15 -17 59 5... 200 10 110 20 200 10 200 130 30 200 100 10 90 190 120 10 200 200 120 90

**T**-matrix

46 -60 -36 11 14

-32 -9 130 105 -37

- 59 -3 27 -12 30
- 50 -71 -24 -56 -40

'Large color change, difficult to compress after DCT'

- To restore, inverse DCT (IDCT) is performed:
- P[x][y] = 0.25 ∑ ∑ C(i) C(j) T[i][j] Cos (2x+1)iπ/16 \* Cos (2y+1)jπ/16

$$x = 0, 1, ..., 7, y = 0, 1, ..., 7$$

# Can write a C-program to apply DCT on a Parray (8\*8) to obtain T-array and also IDCT on T-array to recover P-array.

- Quantization: Provides an way of ignoring small differences in an image that may not be perceptible.
- Another array Q is obtained by dividing each element of T by some number and roundingoff to nearest integer => loss

# T-matrix Q-matrix • 152 0 -48 0 -8... 15 0 -5 0 -1... 0 0 0 0 0 0 0 0 0 0 -48 0 38 0 -3 -5 0 4 0 0 0 0 0 0 0 0 0 0 0 0 0 -8 0 -3 0 13 -1 0 0 0 1

'Divide by 10 and round-off' => 'creates fewer distinct numbers and more consistent pattern'

- Can we recover by multiplying the elements with 10?
- If the loss is minimal, the vision system may not notice.
- Dividing T-elements by the same number is not practical, may result in too much loss.
- Retain the effects of lower spatial frequencies as much as possible – less subtle features noticed if changed

- Upper left elements to be divided with smaller values
- Lower right elements higher spatial frequencies, finer details - to be divided with larger values
- To define a quantization array U, then
   Q[i][j] = Round (T[i][j] ÷ U[i][j]), i = 0, 1, ...7, j = 0, 1, ...7

#### JPEG Compression

U = 1 3 5 7 9... Q= 152 0 -10 0 -1 ...
3 5 7 9 11 0 0 0 0 0
5 7 9 11 13 -10 0 4 0 0
7 9 11 13 15 0 0 0 0 0
9 11 13 15 17 -1 0 0 0 0

'Q can be well-compressed due to redundant elements'

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# JPEG Compression

- Encoding 152 0 -10 0 -1 ...
  - linearize two
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- Row-wise (shorter runs)
- Zigzag (longer runs, higher spatial frequencies are gathered together)

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### JPEG Compression

- Many such 8\*8 arrays, adjacent blocks with little difference => more potential for compression
- JPEG may provide 95% compression (depends on image and quantization array)
- GIF (Graphic Image Format) reduces color to 256. Best suited for a few colors and sharp boundaries (charts, lines). Not good for variations and shading – full color photo

- JPEG compresses still pictures, motion pictures are compressed by MPEG.
- Still picture contains 7, 372,800 bits.
- Compression ratio 20:1 reduces the bits to 368,640
- With 30 images per second, bits to be handled 11,059,200.
- => Huge data transmission if channel is shared

- Based on JPEG compression and relative encoding
- Usually little difference from one to the next frame => suitable for compression
- A completely new scene can not be compressed this way. Also, not suited for moving objects unleashing other objects.
- Three frames: I, P, B.

- I (intra-picture) frame: Just a JPEG encoded image.
- P (predicted) frame: Encoded by computing the differences between a current and a previous frame.
- B (bidirectional) frame: Similar to P-frame except that it is interpolated between previous and future frame.

• I-frames must appear periodically in any frame sequence, otherwise:

 any error introduced in any frame is propagated in all subsequent frames if relative differences are sent

- in broadcast applications, if one tunes-in late, then he has nothing to compare to

Typical sequence:
order in which to transmit: I – P – I
P is sandwiched between groups of B
P is the difference from the previous I-frame
B is interpolated from the nearest I and P
order in which to display I – B – B – P – B – B – I

#### • Coding P:

- Motion compensated Prediction for P-frames
- Divide the image into macro-blocks of 256 pixels (16\*16)
- Chrominance arrays are reduced to groups of 4 pixels (becomes 8\*8 matrix) representing average values => loss, not perceptible

- An algorithm examines each macro-block of Pframe and locates a best matching macroblock in the prior I-frame
- Not necessarily in the same relative position (in the vicinity)
- Once located, the algorithm calculates the difference and motion vector (displacement)
- The above information is encoded and transmitted

 At the decoding end, the motion-vector is used to determine the position and the *difference* is used to reconstruct the macroblock

- MP3 (MPEG Level 3) compression protocol for audio
- Audible range: 20 20 KHz
- PCM uses 16 bit samples @ 44.1 KHz
- 1-sec of PCM audio needs: 16\*44.1\*1000 bits
- For two channel stereo: Twice
- For 1-minute audio: 60 times -> 88 Mbits

- MP3 is based on a 'psychoacoustic model' what we can hear and what we can distinguish
- 'Auditory masking' capture an audio signal, determine what we can not hear, remove those components from the stream, digitize what is left
- 'Sub-band coding' decompose the original signal into non-overlapping frequency ranges, create bit-stream for each band

- Sub-bands are encoded differently:
- Sub-bands with loud signals need good resolution, are coded with more bits
- Near-by sub-bands with weak signals are effectively masked by louder signals, need less resolution, are coded with fewer bits, that results in *compression*