

## LECTURE 22:

## LOSSY COMPRESSION &amp; RATE-DISTORTION

November 29, 2006

- Any lossy compression scheme is based (at least implicitly) on some idea of what counts as “close to the original”.
- This is a question that can only be answered by considering the *users* of the compression program, and what they want.
- For images and audio signals, two fundamental issues are:

– What differences can humans perceive?

For example, to a first approximation, humans perceive only *relative energies* of different frequency bands in audio but not the associated *phases* of sine waves.

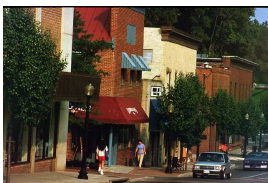


– What differences do humans find annoying or distracting?

Slight changes in colour might be regarded as less important than making a straight line be jagged.



- Many kinds of data — such as images and audio signals — contain “noise” and other information that is not really of interest. Preserving such useless information seems wasteful.
- **A common approach:** *Lossy compression*, for which decompressing a compressed file gives you something *close* to the original, but not necessarily exactly the original.
- We should be able to compress to a smaller file size if we don’t have to reproduce the original exactly.



100% size



5% size

- Suppose the input to the compression program is the sequence  $a_1, a_2, \dots, a_N$ , and the decompression program outputs the sequence  $b_1, b_2, \dots, b_N$ . (The  $a_i$  and the  $b_i$  might come from the same or different alphabets, e.g. before/after quantization.)
- We can measure how close the decompressed output is to the original by its average “distortion”:

$$\bar{d} = \frac{1}{N} \sum_{i=1}^N d(a_i, b_i)$$

$d(a, b)$  is a non-negative *distortion function* measuring how bad it is for a decompressed symbol to be  $b$  if the original was  $a$ .

- **Note:** In practice, the overall distortion might not be a sum of distortions for individual symbols, but we’ll ignore that complication.



- Consider all channels,  $\mathcal{C}$ , with input alphabet  $\{a_i\}$  and output alphabet  $\{b_j\}$ . Given the input probabilities that our source has, we can find for each such channel:
  - Its mutual information,  $I(\mathcal{A}, \mathcal{B})$ .
  - The average distortion between the channel input and the resulting output.
- Shannon proved that the rate distortion function,  $R(D)$ , is equal to the minimum value for  $I(\mathcal{A}, \mathcal{B})$  over all channels whose average distortion is no more than  $D$ .
- For a binary source where 0 has probability  $p_0 \leq 1/2$ , and where distortion is measured by Hamming distance, it turns out that

$$R(D) = \begin{cases} H(p_0) - H(D) & \text{for } 0 \leq D \leq p_0 \\ 0 & \text{for } D > p_0 \end{cases}$$

- As for his noisy coding theorem, Shannon's rate distortion theorem can be proved using codes chosen at random.
- Consider a channel  $\mathcal{C}$  that minimizes  $I(\mathcal{A}, \mathcal{B})$  subject to the distortion between input and output being less than  $D$ . We find the output probabilities for such channel, and then pick codewords at random with symbol probabilities equal to these output probabilities.
- **Encoding procedure:** Find the codeword closest (as measured by distortion) to the actual message; then send an index of that codeword. If we chose  $2^K$  codewords, sending this index will take  $K$  bits, for a rate of  $K/N$ .
- **Decoding procedure:** Output the codeword corresponding to the received index. This is called *vector quantization*.

- Shannon's elegant theory currently plays little role in practical lossy data compression (or the similar task of "vector quantization").
- Instead, various *ad hoc* methods are used.
- Two reasons for this:
  1. Formalizing a suitable distortion function taking account of human perceptual abilities and tolerances is difficult.
  2. The step from the impractical random codes used to prove the rate distortion theorem to a practical method of optimal compression hasn't been achieved.
- Overcoming these issues is a current challenge for research.

