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## **Vector Quantization**

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#### I. Introduction

Vector quantization (VQ) is a lossy data compression method based on the <u>principle of block coding</u>. It is a fixed-to-fixed length algorithm. In the earlier days, the design of a vector quantizer (VQ) is considered to be a challenging problem due to the need for multi-dimensional integration. In 1980, Linde, Buzo, and Gray (LBG) proposed a VQ design algorithm based on a training sequence. The use of a training sequence bypasses the need for multi-dimensional integration. A VQ that is designed using this algorithm are referred to in the literature as an LBG-VQ.

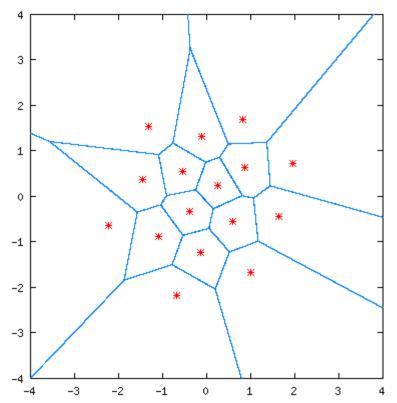
## II. Preliminaries

A VQ is nothing more than an approximator. The idea is similar to that of ``rounding-off'' (say to the nearest integer). An example of a 1-dimensional VQ is shown below:



Here, every number less than -2 are approximated by -3. Every number between -2 and 0 are approximated by -1. Every number between 0 and 2 are approximated by +1. Every number greater than 2 are approximated by +3. Note that the approximate values are uniquely represented by 2 bits. This is a 1-dimensional, 2-bit VQ. It has a rate of 2 bits/dimension.

An example of a 2-dimensional VQ is shown below:



Here, every pair of numbers falling in a particular region are approximated by a red star associated with that region. Note that there are 16 regions and 16 red stars -- each of which can be uniquely represented by 4 bits. Thus, this is a 2-dimensional, 4-bit VQ. Its rate is also 2 bits/dimension.

In the above two examples, the red stars are called *codevectors* and the regions defined by the blue borders are called *encoding regions*. The set of all codevectors is called the *codebook* and the set of all encoding regions is called the *partition* of the space.

## III. Design Problem

The VQ design problem can be stated as follows. Given a vector source with its statistical properties known, given a <u>distortion measure</u>, and given the number of codevectors, find a codebook (the set of all red stars) and a partition (the set of blue lines) which result in the smallest average distortion.

We assume that there is a *training sequence* consisting of M source vectors:

$$\mathcal{T} = \{\mathbf{x}_1, \mathbf{x}_2, \ldots, \mathbf{x}_M\}.$$

This training sequence can be obtained from some large database. For example, if the source is a speech signal, then the training sequence can be obtained by recording several long telephone conversations. M is assumed to be sufficiently large so that all the statistical

properties of the source are captured by the training sequence. We assume that the source vectors are k-dimensional, e.g.,

$$\mathbf{x}_m = (x_{m,1}, x_{m,2}, \dots, x_{m,k}), \quad m = 1, 2, \dots, M.$$

Let N be the number of codevectors and let

$$\mathcal{C} = \{\mathbf{c}_1, \mathbf{c}_2, \ldots, \mathbf{c}_N\},\$$

represents the codebook. Each codevector is k-dimensional, e.g.,

$$\mathbf{c}_n = (c_{n,1}, c_{n,2}, \dots, c_{n,k}), \quad n = 1, 2, \dots, N.$$

Let  $S_n$  be the encoding region associated with codevector  $\mathbf{c}_n$  and let

$$\mathcal{P} = \{S_1, S_2, \ldots, S_N\},\$$

denote the partition of the space. If the source vector  $\mathbf{x}_m$  is in the encoding region  $S_n$ , then its approximation (denoted by  $Q(\mathbf{x}_m)$ ) is  $\mathbf{c}_n$ :

$$Q(\mathbf{x}_m) = \mathbf{c}_n, \quad \text{if } \mathbf{x}_m \in S_n.$$

Assuming a squared-error distortion measure, the average distortion is given by:

$$D_{ave} = \frac{1}{Mk} \sum_{m=1}^{M} ||\mathbf{x}_m - Q(\mathbf{x}_m)||^2,$$

where  $||\mathbf{e}||^2 = e_1^2 + e_2^2 + \ldots + e_k^2$ . The design problem can be succinctly stated as follows: Given  $\mathcal{T}$  and N, find  $\mathcal{C}$  and  $\mathcal{P}$  such that  $D_{ave}$  is minimized.

## IV. Optimality Criteria

If C and P are a solution to the above minimization problem, then it must satisfied the following two criteria.

Nearest Neighbor Condition:

$$S_n = \{\mathbf{x} : ||\mathbf{x} - \mathbf{c}_n||^2 \le ||\mathbf{x} - \mathbf{c}_{n'}||^2 \ \forall n' = 1, 2, \dots, N\}$$

This condition says that the encoding region  $S_n$  should consists of all vectors that are closer to  $\mathbf{c}_n$  than any of the other codevectors. For those vectors lying on the boundary (blue lines), any tie-breaking procedure will do.

• Centroid Condition:

$$\mathbf{c}_n = \frac{\sum_{\mathbf{x}_m \in S_n} \mathbf{x}_m}{\sum_{\mathbf{x}_m \in S_n} 1} \quad n = 1, 2, \dots, N$$

This condition says that the codevector  $\mathbf{c}_n$  should be average of all those training vectors that are in encoding region  $S_n$ . In implementation, one should ensure that at least one training vector belongs to each encoding region (so that the denominator in the above equation is never 0).

## V. LBG Design Algorithm

The LBG VQ design algorithm is an iterative algorithm which alternatively solves the above two optimality criteria. The algorithm requires an initial codebook  $C^{(0)}$ . This initial codebook

is obtained by the *splitting* method. In this method, an initial codevector is set as the average of the entire training sequence. This codevector is then split into two. The iterative algorithm is run with these two vectors as the initial codebook. The final two codevectors are splitted into four and the process is repeated until the desired number of codevectors is obtained. The algorithm is summarized below.

#### LBG Design Algorithm

1. Given Fixed to be a "small" number.

$$T$$
  $\epsilon > 0$ 

2. Let N = 1 and

$$\mathbf{c}_1^* = \frac{1}{M} \sum_{m=1}^M \mathbf{x}_m.$$

Calculate

$$D_{ave}^* = \frac{1}{Mk} \sum_{m=1}^{M} ||\mathbf{x}_m - \mathbf{c}_1^*||^2.$$

3. **Splitting:** For  $i = 1, 2, \dots, N$ , set

$$\mathbf{c}_{i}^{(0)} = (1+\epsilon)\mathbf{c}_{i}^{*},$$
  

$$\mathbf{c}_{N+i}^{(0)} = (1-\epsilon)\mathbf{c}_{i}^{*}.$$

Set N=2N.

4. Iteration: Let  $D_{ave}^{(0)} = D_{ave}^*$ . Set the iteration index i = 0.

i. For  $m=1,2,\ldots,M$  , find the minimum value of

$$||\mathbf{x}_m - \mathbf{c}_n^{(i)}||^2$$

over all  $n=1,2,\ldots,N$  . Let  $n^*$  be the index which achieves the minimum. Set

$$Q(\mathbf{x}_m) = \mathbf{c}_{n^*}^{(i)}.$$

ii. For  $n=1,2,\ldots,N$  , update the codevector

$$\mathbf{c}_n^{(i+1)} = \frac{\sum_{Q(\mathbf{x}_m) = \mathbf{c}_n^{(i)}} \mathbf{x}_m}{\sum_{Q(\mathbf{x}_m) = \mathbf{c}_n^{(i)}} 1}$$

iii. Set i = i + 1.

iv. Calculate

$$D_{ave}^{(i)} = \frac{1}{Mk} \sum_{m=1}^{M} ||\mathbf{x}_m - Q(\mathbf{x}_m)||^2.$$

v. If  $(D_{ave}^{(i-1)}-D_{ave}^{(i)})/D_{ave}^{(i-1)}>\epsilon$  , go back to Step (i).

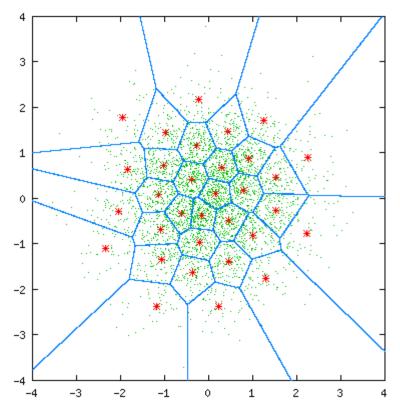
vi. Set  $D_{ave}^* = D_{ave}^{(i)}$ . For  $n = 1, 2, \dots, N$ , set

$$\mathbf{c}_n^* = \mathbf{c}_n^{(i)}$$

as the final codevectors.

5. Repeat Steps 3 and 4 until the desired number of codevectors is obtained.

## VI. Two-Dimensional Animation



Click on the figure above to begin the animation

- If the animation appears to be stuck, try moving up or down the page in your browser.
- The source for the above is a memoryless Gaussian source with zero-mean and unit variance.
- The tiny green dots are training vectors --- there are 4096 of them.
- The LBG design algorithm is run with  $\epsilon = 0.001$ .
- The algorithm guarantees a locally optimal solution.
- The size of the training sequence should be sufficiently large. It is recommended that  $M \geq 1000N$  .

## VI. Performance

The performance of VQ are typically given in terms of the signal-to-distortion ratio (SDR):

$$SDR = 10 \log_{10} \frac{\sigma^2}{D_{ave}} \text{ (in dB),}$$

where  $\sigma^2$  is the variance of the source and  $D_{ave}$  is the average squared-error distortion. The

higher the SDR the better the performance. The following tables show the performance of the LBG-VQ for the memoryless Gaussian source and the first-order Gauss-Markov source with correlation coefficient 0.9. Comparisons are made with the optimal performance theoretically attainable,  $SDR_{opt}$ , which is obtained by evaluating the <u>rate-distortion function</u>.

Rate	SDR (in dB)								$SDR_{opt}$
(bits/dimension)	n = 1	n = 2	n = 3	n = 4	n = 5	n = 6	n = 8	n = 10	$n=\infty$
1	4.4	4.4	4.5	4.7	4.8	4.8	4.9	5.0	6.0
2	9.3	9.6	9.9	10.2	10.3				12.0

3	14.6	15.3	15.7						18.1
4	20.2	21.1							24.1
5	26.0	27.0							30.1
Memoryless Gaussian Source									
	_	_	_	_	_	_	_	_	_
Rate	SDR (in dB)							$SDR_{opt}$	
(bits/dimension	n = 1	n=2	n = 3	n = 4	n = 5	n = 6	n = 8	n = 10	$n=\infty$
1	4.4	7.8	9.4	10.2	10.7	11.0	11.4	11.6	13.2
2	9.3	13.6	15.0	15.8	16.2				19.3
3	14.6	19.0	20.6						25.3
4	20.2	24.8							31.3
5	26.0	30.7							37.3
First-Order Gauss-Markov Source with Correlation 0.9									

## **VIII. References**

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