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Source Coding

Tutorial Problems

WS 2020/2021

Hints:

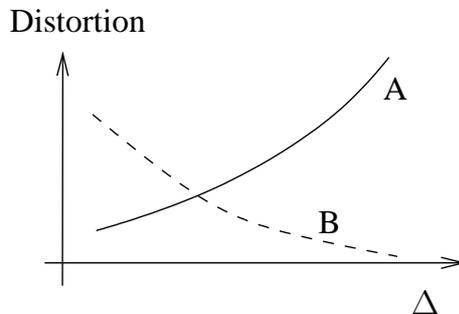
Parts of a problem that are marked by " \implies " can be solved without the solutions of previous parts.

This material is supplementary but not relevant for the exam.

Problem 1

Consider uniform quantisation (with step-size Δ) of a source signal with unlimited amplitude range.

In the figure below both the characteristics of the granular distortion and the overload distortion are depicted versus the step-size Δ of the quantiser.



- (a) Label the curves appropriately and give reasons for your answer.
- ⇒ (b) State, for a uniform quantiser with mid-riser characteristic, the reproducer values and the decision boundaries as functions of Δ and the rate R .
- ⇒ (c) Give the general formula for the power of the quantiser error and state how in principle the optimal value for Δ can be found (no calculations!).

Problem 2

In this problem we will consider high-rate R scalar uniform quantisation of a Laplacian source with the PDF

$$p(x) = \frac{1}{2} e^{-|x|}.$$

- (a) Show how to compute the factor $K = \frac{P_x}{x_{\max}^2}$ (see lecture notes). Give the exact value for K dependent on the probability $p_1 \doteq \Pr\{|x| > x_{\max}\}$. Calculate the value of x_{\max} for $p_1 = 0.001$.
- (b) Compute the overload distortion for $p_1 = 0.001$. For this, assume approximately that the reproducer values that are used to quantise the overload regions are $\hat{x}_0 = -x_{\max}$ and $\hat{x}_{2R-1} = x_{\max}$.
Is the true value of the overload-distortion over- or under-estimated by the approximation? (Why?)
- (c) Is it justified to always neglect the overload distortion for $p_1 = 0.001$, or is that a matter of rate? To clarify this issue, assume that the granular distortion equals $P_Q = \Delta^2/12$, write the SNR in the form

$$\frac{SNR}{\text{dB}} = 6.02 \cdot R + 10 \log_{10}(3K) - 10 \log_{10}(1 + c).$$

How large is $10 \log_{10}(1 + c)$ for $p_1 = 0.001$ and $R = 5$?

Problem 3

In this problem we will compute the SNR that results from logarithmic quantisation (A-law characteristic); the result is given in the lecture notes. We will assume a symmetric PDF $p(x)$ for the source signal samples.

- (a) For the contribution to the total noise power that results from the i -th quantiser interval we obtain approximately

$$P_{Q_i} \approx \frac{\Delta x_i^2}{12} \int_{\hat{x}_i - \Delta x_i/2}^{\hat{x}_i + \Delta x_i/2} p(x) dx = \frac{\Delta x_i^2}{12} p_i ,$$

where \hat{x}_i is the i -th reproducer value and Δx_i is the size of the corresponding quantiser interval, which is *not* equal for all i .

Which approximation is used? Which quantity is described by p_i ?

- (b) The total quantiser noise power P_Q is given by the sum of all interval noise powers P_{Q_i} :

$$P_Q = \sum_{i=0}^{L-1} P_{Q_i} = 2 \sum_{i=L/2}^{L-1} P_{Q_i} \approx 2 \sum_{i=L/2}^{L-1} \frac{\Delta x_i^2}{12} p_i$$

Explain the second equality.

- (c) The width of a quantiser interval follows from the compressor characteristic $y = g(x)$:

$$\Delta x_i \approx \frac{\Delta y}{g'(\hat{x}_i)}$$

Explain this by elementary differential calculus.

- (d) In the derivation of the A-law compressor we required the interval-width (in the x -domain) to be proportional to the signal amplitude. From this and from (c) we obtain

$$\frac{1}{g'(x)} \stackrel{!}{=} c x \quad \text{with } c \text{ a constant}$$

Determine an expression for Δx_i from (c) and the expression above.

- (e) For the total noise power we finally obtain

$$P_Q \approx \frac{\Delta y^2}{12} c^2 P_x ,$$

where P_x is the power of the input signal.

Derive this formula.

- (f) The compressed samples y are uniformly quantised by R bits per sample. Assume that $x_{\max} = y_{\max} = 1$ (which means no loss of generality) and insert a relation for Δy into the solution of (e).
- (g) Compute the SNR in dB according to

$$\frac{SNR}{\text{dB}} = 10 \log_{10} \left(\frac{P_x}{P_Q} \right)$$

The result is

$$\frac{SNR}{\text{dB}} = R \cdot 20 \log_{10}(2) + 10 \log_{10}(3) - 20 \log_{10}(c)$$

Give an interpretation of this result. Does it depend on the signal power?

- (h) Compute the factor c for the A-law case in the amplitude range $\frac{1}{A} \leq |x| \leq 1$ (Hint: $1/g'(x) = c x$)
- (i) As a result we obtain

$$\frac{SNR}{\text{dB}} = R \cdot 6.02 - 9.99 \quad \text{for} \quad \frac{1}{A} \leq |x| \leq 1$$

Compare this result with conventional uniform quantisation.

- (j) For small signal amplitudes (we implicitly assumed large amplitudes above) the logarithmic quantiser behaves like a uniform quantiser (with a different bit rate).
Why and for which amplitudes of the input signal x is this the case?
- (k) Compute the gradient of the A-law characteristic for small values of x .
- (l) Determine the effective interval size of the logarithmic quantiser (A-law) for small values of x .
- (m) Give the resulting SNR and compare it with the result of a conventional uniform quantiser with the same bit rate R .

Problem 4

Consider the probability density function (PDF)

$$p(x) = \begin{cases} |x|, & -1 \leq x \leq 1 \\ 0, & \text{else} \end{cases}.$$

Independent realisations of the random variable X with the PDF $p(x)$ are to be scalar quantised by $R = 2$ bits per sample.

- (a) Sketch the PDF $p(x)$ and determine its mean μ_x , its variance σ_x^2 , and the power P_x of a random process that has independent realisations, each with the PDF $p(x)$.

- \Rightarrow (b) Give the quantiser decision boundaries $g_i, i = 0, 1, \dots, 4$, and the reproducer values $\hat{x}_i, i = 0, 1, \dots, 3$, for uniform quantisation (mid-riser characteristic, $x_{\max} = 1$).

Sketch the quantiser characteristic; label the axes.

- (c) Give the general formula to compute the power P_Q of the quantiser noise (dependent on the reproducer values and the decision boundaries)?
- (d) Show that in our case

$$P_Q = \frac{3}{4} \hat{x}_0^2 + \frac{7}{6} \hat{x}_0 + \frac{1}{4} \hat{x}_1^2 + \frac{1}{6} \hat{x}_1 + \frac{1}{2}$$

holds. For this, simplify the general expression for P_Q in (c) by use of the symmetries of the PDF and insert numerical values only for the decision boundaries g_i .

- (e) Compute the value of P_Q and give the quantiser SNR in dB.

In what follows the quantiser decision boundaries g_i will be adopted from (b), but the reproducer values $\hat{x}_i, i = 0, 1, \dots, 3$, shall be selected such that the power P_Q of the quantiser error is minimised.

- \Rightarrow (f) Compute the new reproducer values $\hat{x}'_i, i = 0, 1, \dots, 3$.
- (g) Sketch the new quantiser characteristic and label the axes with variables and numerical values.
- (h) Calculate the new power P'_Q of the quantiser error and the new SNR' in dB.
- (i) Compare the SNR-values in (e) and (h) and explain potential differences.
- (j) How could the quantiser performance be further improved?

Problem 5

We want to scalar quantise independent Gaussian samples X with the PDF

$$p(x) = \frac{1}{\sqrt{2\pi}\sigma} \exp\left(-\frac{1}{2\sigma^2}x^2\right).$$

Further, we assume that the rate R (bits per sample) is high.

- (a) Determine the signal power P_x and give the quantiser SNR of a scalar uniform quantiser for an overload probability of $P_o = \Pr\{|x| > x_{\max}\} = 0.001$ (neglect the overload distortion in the SNR computation).
- \implies (b) How large is the SNR in dB of the best possible scalar *fixed-rate* quantiser?
- \implies (c) Which maximum SNR can be obtained by the best scalar *variable-rate* scalar quantiser?
- \implies (d) What is the maximum SNR in dB of *any* quantisation scheme without *any* restrictions.

In what follows we will assume that the samples X are quantised by a companding scheme, which uses the compressor-function

$$y = g(x) = \alpha \cdot \frac{2}{\pi} \cdot \arctan(\beta x) \quad \text{with} \quad \alpha, \beta > 0.$$

In the y -domain, the samples are uniformly quantised.

- \implies (e) Into which interval (y_{\min}, y_{\max}) is the support $(-\infty, +\infty)$ of the PDF $p(x)$ mapped by the compressor-characteristic $g(x)$?
- (f) Compute the quantiser distortion $P_Q(R)$ as a function of α and β ? Do both variables influence P_Q ?

Hint:

$$\int_0^\infty x^n \cdot e^{-ax^2} dx = \frac{1 \cdot 3 \cdot \dots \cdot (2k-1)\sqrt{\pi}}{2^{k+1} \cdot a^{k+1/2}} \quad \text{for} \quad n = 2k, k = 1, 2, \dots$$

- (g) Determine the variables α and β such that the quantiser distortion computed in (f) is minimised and calculate the optimal quantiser SNR in dB.

Now we assume a fixed *low* rate of $R = 1$ bit / sample, and we want do design the best possible fixed-rate scalar quantiser; we assume, however, symmetric reproducer values ($\hat{x}_0 = -\hat{x}_1$).

- \implies (h) Determine the decision boundaries g_0, g_1, g_2 (use the symmetry of the source PDF).
- (i) Calculate the optimal reproducer values.
- (j) Calculate the quantiser distortion P_Q and the corresponding SNR in dB.
- (k) Discuss the difference in SNR of the results from (j) and (b).

Problem 6

In the figure (see overleaf), code-vectors of a two-dimensional vector quantiser are given.

(a) How many bits are required to transmit the code-vectors by a fixed bit rate per index? Are the bits efficiently used (in terms of source coding)?

⇒ (b) Insert the Voronoi-regions into the figure; assume that the mean squared error is used as a distance measure.

Hint: consider first, how the partitioning of the space looks like, if only two code-vectors are used.

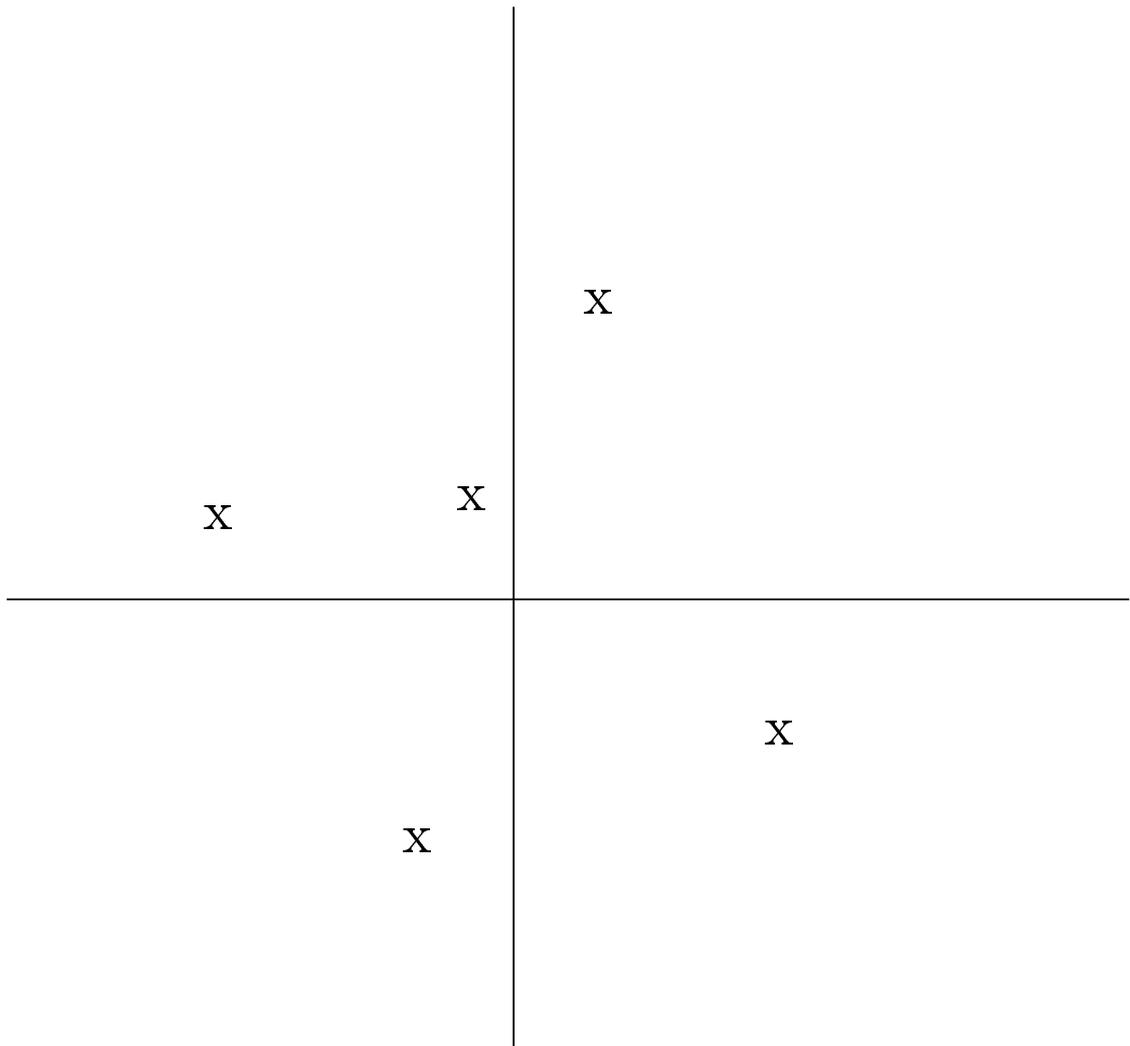
⇒ (c) Is there any potential gain from using bits inefficiently from a source-coding point-of-view?

Hint: think about bit errors.

⇒ (d) Assume the codevectors are uniformly distributed. How large is their entropy?

(e) Construct a Huffman code for the codevectors. How large is the average word-length? Compare the result with the entropy and with a fixed-rate index allocation.

(Problem 6)



Problem 7

In this problem, we will illustrate how the LBG algorithm for vector quantiser codebook training works.

In the figure entitled “1. Iteration,” initial code-vectors (marked by “x”) are given; the mid-points of lines connecting them are marked by “o”. Moreover, data points are also given by the dots.

We use the mean squared error (MSE) as a distance measure.

- (a) Show analytically that the centroids are given by the average of the data-vectors that are allocated to a quantiser region (Voronoi region) by the first step of the iterative training procedure.
- ⇒ (b) Add the Voronoi regions for the given code-vectors into the figure entitled “1. Iteration.”
- (c) Determine (graphical approximation, no calculations) the location of the “new” code-vectors in each Voronoi region.
- (d) Copy the new code-vectors resulting from (c) into the figure entitled “2. Iteration” and determine the Voronoi regions for these code-vectors.
- (e) Determine the location of the code-vectors after the second iteration.

Now we consider vector quantisation with the following modified distance measure:

$$d(\underline{x}_k, \underline{y}_i) = f(\underline{x}_k) \cdot \sum_{j=0}^{N-1} (x_k(j) - y_i(j))^2 \quad \text{with} \quad f(\underline{x}_k) > 0$$

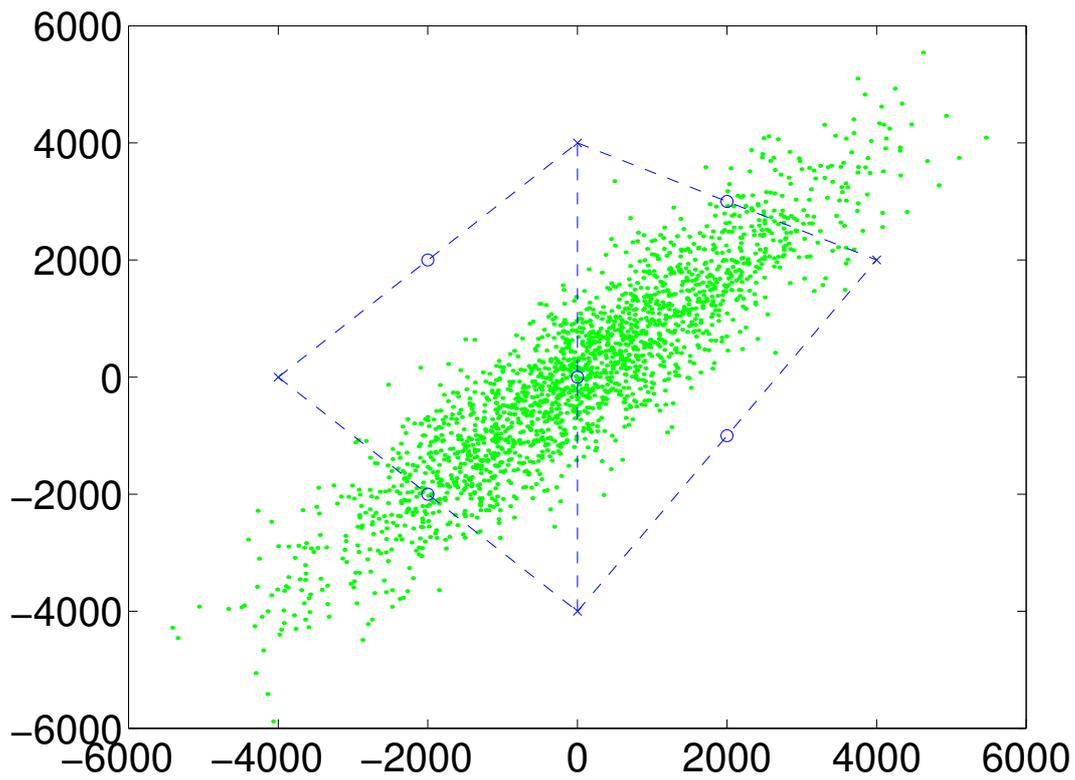
The data vector to be quantised is given by $\underline{x}_k = \{x_k(j), j = 0, 1, \dots, N - 1\}$, the codevector with the index i is given by $\underline{y}_i = \{y_i(j), j = 0, 1, \dots, N - 1\}$.

- ⇒ (f) Assume a codebook with the codevectors \underline{y}_i is given and some data vector \underline{x}_k is to be quantised: will the use of the modified distance measure make a difference compared with the conventional MSE distance measure?
- ⇒ (g) Compute, for a general function $f(\underline{x}_k)$, the “new” centroids of the partition regions in Step 4 of the LBG algorithm.
- (h) We now assume

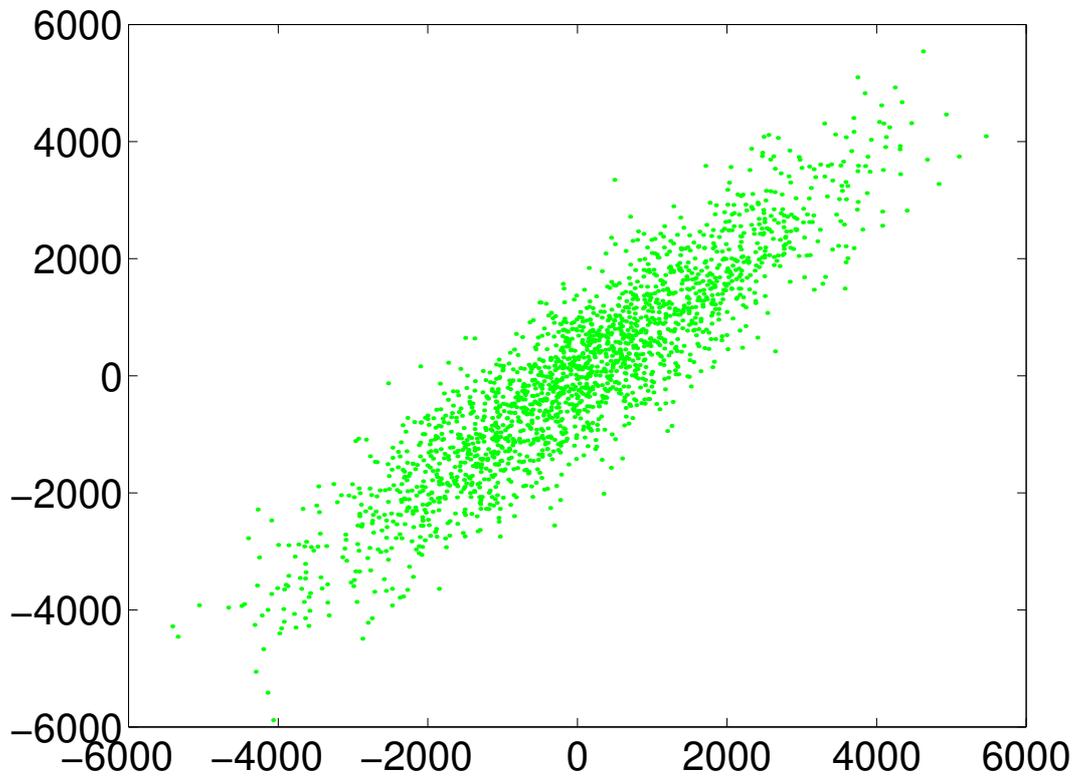
$$f(\underline{x}_k) = \frac{1}{\|\underline{x}_k\|^2} = \frac{1}{\sum_{j=0}^{N-1} x_k^2(j)} .$$

Discuss the influence of the distance measure for the location of the code-vectors that result from LBG codebook training.

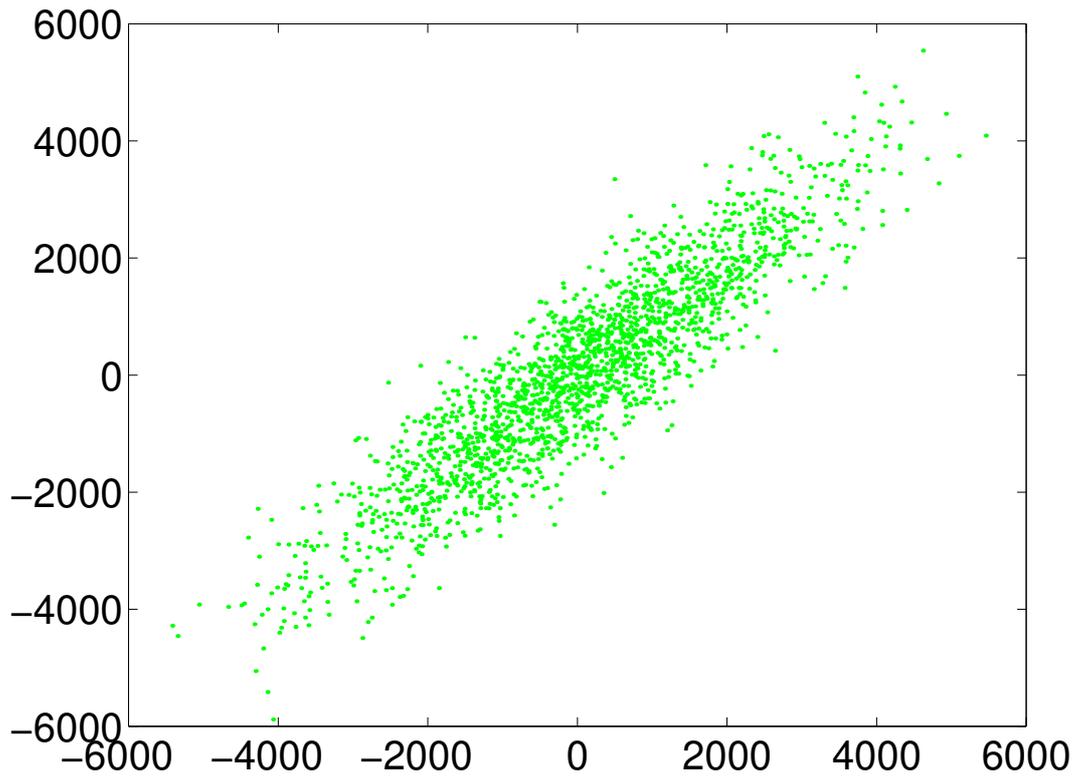
1. Iteration



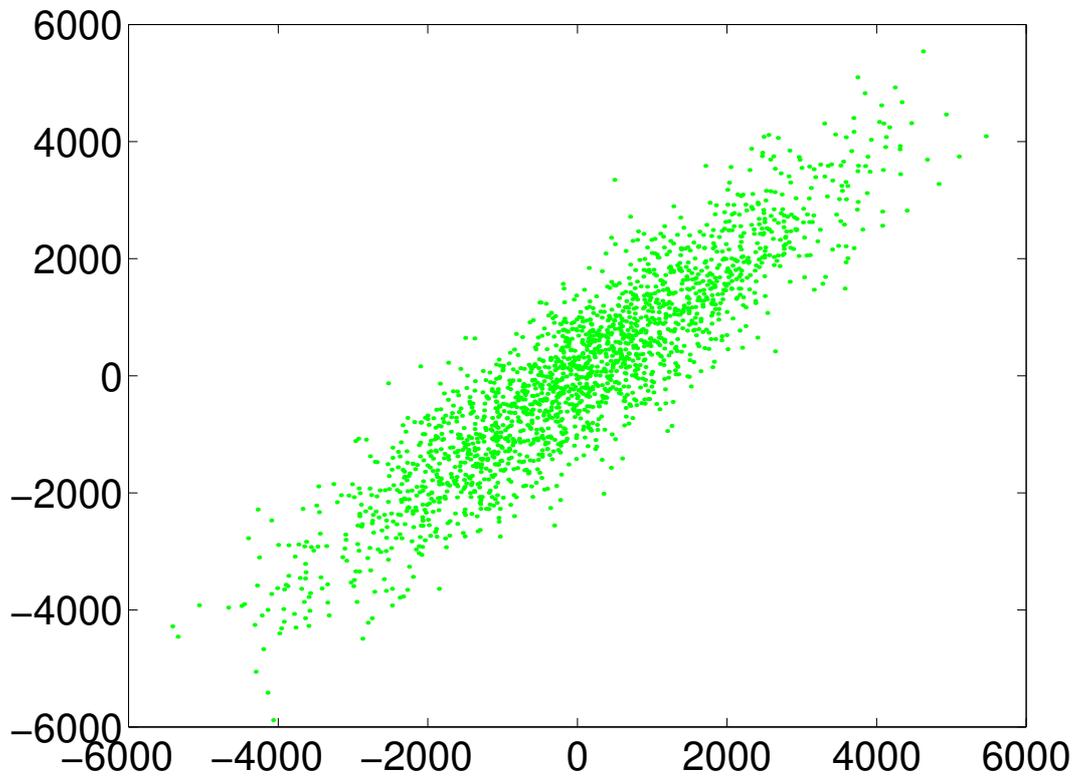
1. Iteration



2. Iteration



2. Iteration



Problem 8

In the lectures, we found the rate-distortion function for a memoryless Gaussian source and squared error distortion.

There is one more case in which an analytical result is known: the binary Bernoulli(p) source¹ with an expected portion of errors in the sequence less than or equal to D . The corresponding per-letter distortion measure (the Hamming distortion measure) is simply “1” if $X_i \neq \tilde{X}_i$ and “0” if $X_i = \tilde{X}_i$, where X_i is a letter from the source sequence and \tilde{X}_i its reproduction.

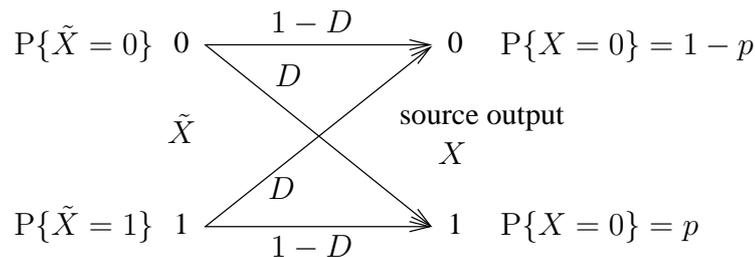
In this problem we will prove the rate-distortion function

$$R(D) = \begin{cases} H_2(p) - H_2(D), & 0 \leq D \leq \min\{p, 1-p\} \\ 0, & D > \min\{p, 1-p\} \end{cases}$$

for the Bernoulli source and the Hamming (per-letter) distortion measure. The quantity $H_2(p) = -p \cdot \log_2(p) - (1-p) \cdot \log_2(1-p)$ is the so-called binary entropy function.

The steps of the proof are similar to the Gaussian case.

- (a) State the general expression for the (information) rate-distortion function for sources with letter alphabets of limited size.
- (b) Find a sequence of inequalities (as in the Gaussian case) to derive a lower bound for the mutual information $I(X; \tilde{X})$.
Hint: consider the conditional entropy of $X \oplus \tilde{X}$ given \tilde{X} , where “ \oplus ” denotes modulo-2 addition.
- (c) In the second step of the proof, consider the following system:



Compute the probabilities $\Pr\{\tilde{X} = 1\} = 1 - \Pr\{\tilde{X} = 0\}$.

Prove² that the system achieves the lower bound derived in (c).

¹Such a source emits a “1” with probability p and a “0” with probability $1 - p$.

²Only if really necessary, consider the book by Cover, Thomas.

Problem 9

We want to show that the inverse of the spectral flatness measure for a Gauss-Markov process, which is generated by linear filtering of a Gaussian source with the transfer function

$$H(z) = \left(1 + \sum_{i=1}^{N_z} b_i \cdot z^{-i} \right) / \left(1 + \sum_{i=1}^{N_p} a_i \cdot z^{-i} \right),$$

is given by

$$\frac{1}{\gamma_x^2} = \frac{\sigma_x^2}{\sigma_r^2} = \frac{1}{2\pi} \int_{-\pi}^{+\pi} |H(\Omega)|^2 d\Omega = \sum_{k=0}^{\infty} h_0^2(k)$$

where $h_0(k)$ is the discrete-time impulse response of the filter.

- What is the definition of the spectral flatness measure?
- Factorise the polynomials in the numerator and the denominator of $H(z)$ assuming general poles $z_{\infty,i}$ and zeros $z_{0,i}$, but for simplicity assume that all poles and zeros are real numbers.
- Write the square of the magnitude of a complex number by an alternative expression (that does not use “ $|\cdot|$ ”).
- Use the hint below to compute the spectral flatness measure. Consider that the filter must be stable and have minimum phase (what does that mean for the poles and zeros?).
- Explain the rightmost equality in the formula for the spectral flatness measure stated above.

Hint:

$$\int_0^{\pi} \log_e(a^2 - 2ab \cos(x) + b^2) dx = \begin{cases} 2\pi \log_e(a), & a \geq |b| > 0 \\ 2\pi \log_e(b), & b \geq |a| > 0 \end{cases}$$

Problem 10

We are given a memoryless discrete-time Gaussian source which emits the samples $r(k)$. The latter are filtered by a discrete-time causal filter with the transfer function

$$H(z) = \frac{z^2}{(z-a)(z-b)} \quad \text{with} \quad a = \frac{1}{4}, \quad b = \frac{3}{4}.$$

The resulting output signal is denoted by $x(k)$.

In what follows, we will assume high-rate coding.

- (a) What is the maximum SNR we can obtain by coding the source with the samples $r(k)$ at some rate R with a coding scheme without any restrictions?
- (b) Is the result (a) also true for low rates? Why?
- (c) Compute the impulse response $h_0(k)$ of the filter.
- (d) Determine the inverse $1/\gamma^2$ of the spectral flatness measure of $x(k)$.
- (e) Which maximum SNR (in dB) can be achieved by any coding scheme (no restrictions) if the signal $x(k)$ is coded at some high rate R ?
- (f) Which PDF have the individual samples $x(k)$?
- (g) Which SNR in dB would result from a scalar fixed-rate optimal quantisation of $x(k)$ without use of the signal correlation for coding?
- (h) Which maximum SNR in dB would result from a scalar fixed-rate optimal quantisation if we use a certain coding system that exploits the signal correlation of $x(k)$? What are the names of such a system?

Hints:

$$\sum_{i=0}^{\infty} q^i = \frac{1}{1-q} \quad \text{for} \quad 0 < |q| < 1$$

$$Z\{\alpha^k \gamma_{-1}(k)\} = \frac{z}{z-\alpha} \quad \text{with} \quad \gamma_{-1}(k) = \begin{cases} 1, & k \geq 0 \\ 0, & k < 0 \end{cases}$$