## **Digital Signal Processing - DSP**

(PDS - Processamento Digital de Sinais) Instituto Superior Técnico - 2º semestre 2008/2009 Jorge Salvador Marques

## **Random Signals**

Most signals observed in nature, in society and in economy are random signals. This means that everytime we perform a new observation we obtain a different signal. Examples of random signals are: speech, audio, ECG, EEG, economic series.

We will consider two signals in this work: Gaussial signals and speech signals. Gaussian signals can be automatically generated in a computer using a random number generator. The random generator produces a sequence of independent realizations of a Gaussian variable with distribution N(0, 1). The autocorrelation of this sequence is  $r(k) = \delta(k)$  since different samples are uncorrelated. We can collect several variables in a single vector x with normal distribution N(0, I). Dependence among the variables can be created by performing a linear transformation y = Ax.

Real signals are more complex. For example, speech signals have a rich time structure. They exhibit short range dependence among samples separated by less than 2 ms and long range dependence (3-15 ms) due to the periodic vibration of the vocal folds.

## **Gaussian Signals**

- 1. Generate 1000 realizations of a Gaussian random vector  $x(n) \sim N(0, I)$  with dimension d = 2. Visualize the data. Suggestion: use Matlab command randn.
- 2. Second order statistics: determine the mean and covariance matrix of the data using the following estimators

$$\hat{\mu} = \frac{1}{N} \sum_{n=1}^{N} x(n) \qquad \hat{R} = \frac{1}{N} \sum_{n=1}^{N} (x(n) - \hat{\mu})^T (x(n) - \hat{\mu})$$
(1)

Compare with the true values. Suggestion: use the commands mean e cov.

- 3. Determine the 2D histogram of vector x splitting the plane  $\mathbb{R}^2$  into cells of amplitude  $0.5\times0.5.$
- 4. Transform the data using the affine transform y = Ax + b with  $A = \begin{bmatrix} 1 & .8 \\ .5 & 1 \end{bmatrix}$   $b = \begin{bmatrix} 1 \\ 2 \end{bmatrix}$ . Visualize the transformed data. What differences do you notice.
- 5. Determine the mean and covariance of the vector y. What are the theoretical values you expect?

## Speech signal

- 6. Read the signal speech wav and visualize it. Find quasi periodic structure due to the vibration of the vocal folds.
- 7. Forget the time dependence and compute the histogram of speech. Does it look Gaussian? compare with a Gaussian pdf with the same mean and variance.
- 8. Select an interval in which the speech signal is quasi periodic. Determine the autocorrelation function  $r(k) = \sum_{n=0}^{N-1} x(n)x(n+k)$  and plot it. Try to identify the short range dependence and the long range dependence.
- 9. How could you use the autocorrelation function to determine the period?