

TSKS04 Digital Communication

Continuation Course

Lecture 9

Linear Equalization, Block codes, Convolutional codes

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Linear Equalization

Time-discrete model

$$\mathbf{x}[n] = \mathbf{U} \cdot \mathbf{B}[n] + \mathbf{w}[n]$$

Received signal vector $L \times 1$ Matrix $L \times K$ Noise vector $L \times 1$

Symbol vector $K \times 1$

$$\mathbf{B}[n] = (b[n - k_1], \dots, b[n], \dots, b[n + k_2])^T$$
$$K = k_1 + k_2 + 1$$

Linear equalization

- Take inner product between $\mathbf{x}[n]$ and some vector \mathbf{c}
- Select the $L \times 1$ vector \mathbf{c} so that $\mathbf{c}^H \mathbf{x}[n]$ gives good estimate of $b[n]$

Different Linear Equalization Choices (1/2)

- Matched filter (MF) equalization
 - Maximize desired signal: $\mathbf{c}^H \mathbf{u}_0$
 - Fix $\mathbf{c}^H \mathbf{u}_0 = 1$: $\mathbf{c} = \mathbf{u}_0 / \|\mathbf{u}_0\|^2$
- Zero-forcing (ZF) equalization
 - Force ISI to be zero: $\mathbf{c}^H \mathbf{U} = \underbrace{(0 \dots 0)}_{k_1} 1 \underbrace{0 \dots 0}_{k_2} = \mathbf{e}^T$
 - Achieved by $\mathbf{c}_{ZF} = \mathbf{U}(\mathbf{U}^H \mathbf{U})^{-1} \mathbf{e}$

Different Linear Equalization Choices (2/2)

- Comparison
 - MF: Maximize desired signal
 - ZF: Remove inter-symbol interference
 - What one to choose?
- Minimum mean squared error (MMSE) equalization
 - Balance between strong signal and little interference
 - Mean squared error:

$$J(\mathbf{c}) = E[|\mathbf{c}^H \mathbf{x}[n] - b[n]|^2]$$
$$\mathbf{c}_{MMSE} = \sigma_b^2 (\sigma_b^2 \mathbf{U}\mathbf{U}^H + \mathbf{C}_w)^{-1} \mathbf{u}_0$$

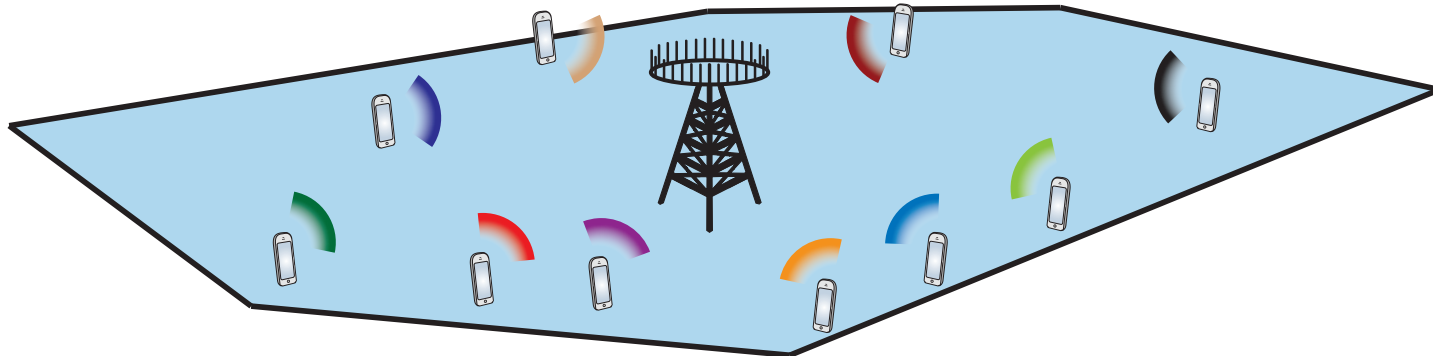
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Noise covariance matrix

Outlook: Equalization of Multi-user Channels

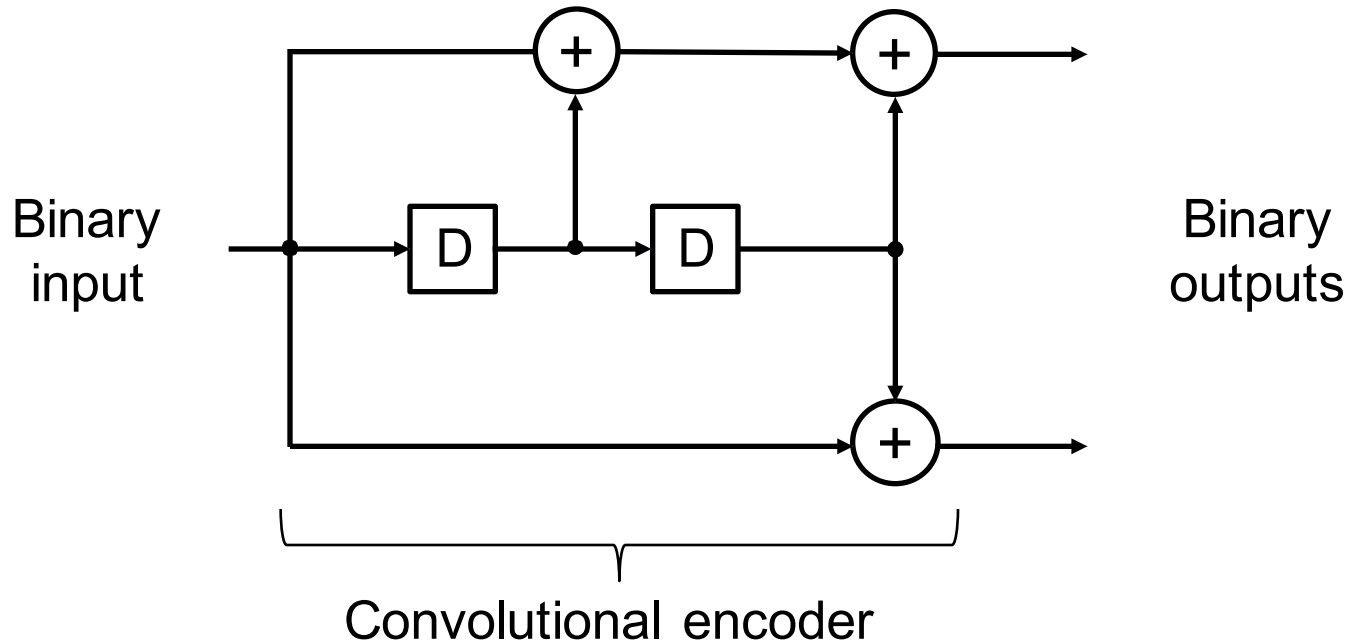
$$\mathbf{x}[n] = \mathbf{U} \cdot \mathbf{B}[n] + \mathbf{w}[n]$$

Received signal vector $L \times 1$ Matrix $L \times K$ Symbol vector $K \times 1$ Noise vector $L \times 1$

- Two scenarios, same equations:
 - Single-user channel with ISI: K symbols and L samples
 - Multi-user channel without ISI: K users and L receive antennas

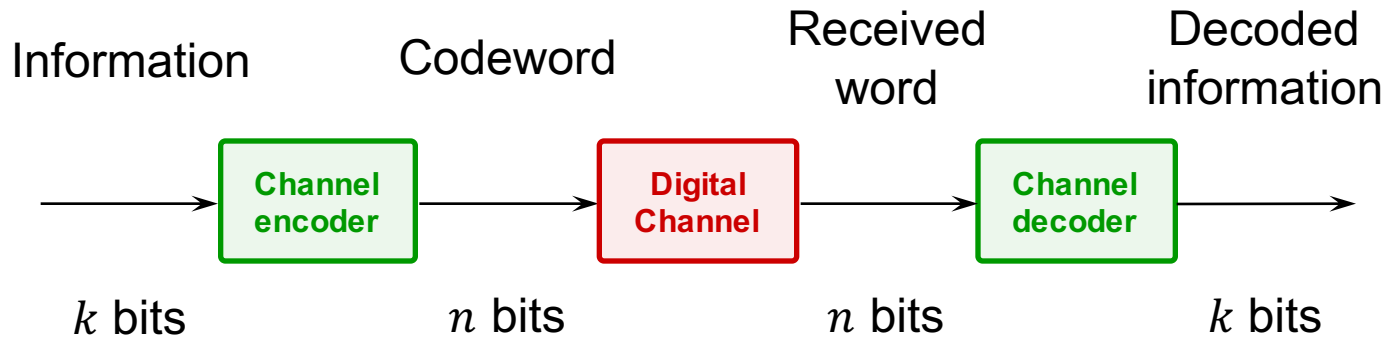


Introduction: Convolutional Codes

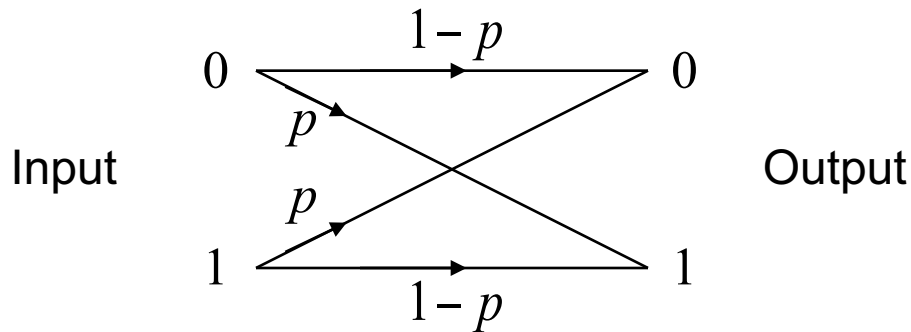


- Encoder: FIR filter, convolution
- State diagram, trellis
- Decoding: Viterbi algorithm

Repetition: Error Control Coding



- Simple channel model: Bits in, bits out
 - Example: Binary symmetric channel



p = Bit error probability

Repetition: Binary Linear Codes $[n, k, d]$

A vector space expressed in a basis

$$\mathcal{C} = \{\bar{m}G \mid \forall \bar{m} \in \mathbb{F}_2^k\}$$

Generator matrix ($k \times n$),
linearly independent rows.

... the nullspace of a matrix

$$\mathcal{C} = \{\bar{c} \in \mathbb{F}_2^n : H\bar{c}^T = \bar{0}\}$$

Parity check matrix ($(n - k) \times n$),
linearly independent rows.

$$HG^T = 0$$

Length: n , # columns in G or H

Dimension: k , # rows in G .

Minimum distance, d

Smallest Hamming distance between different codewords.

Smallest Hamming weight of non-zero codewords.

Smallest number of linearly dependent columns in H .

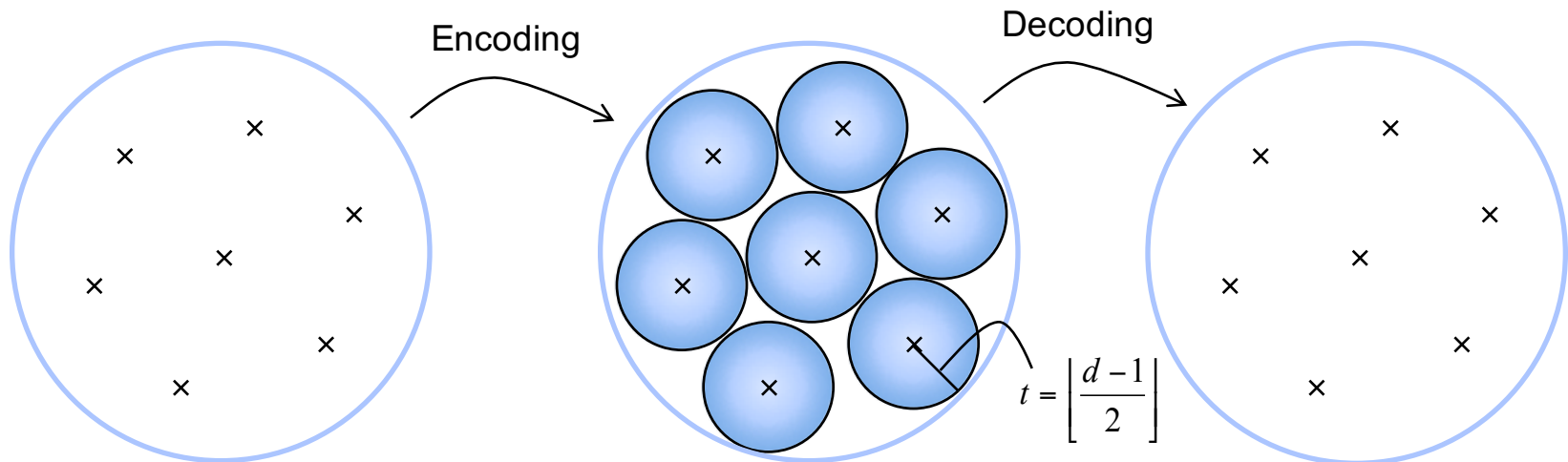
Repetition: Decoding of block codes

- Syndrome decoding

- Parity check matrix: $H\bar{c}^T = \bar{0}$

- Syndrome:
$$\bar{s} = H\bar{x}^T = H \cdot (\bar{c} + \bar{e})^T = \underbrace{H\bar{c}^T}_{=\bar{0}} + H\bar{e}^T = H\bar{e}^T$$

- **Syndrome decoding:** Find smallest set of columns in H that sum up to the syndrome



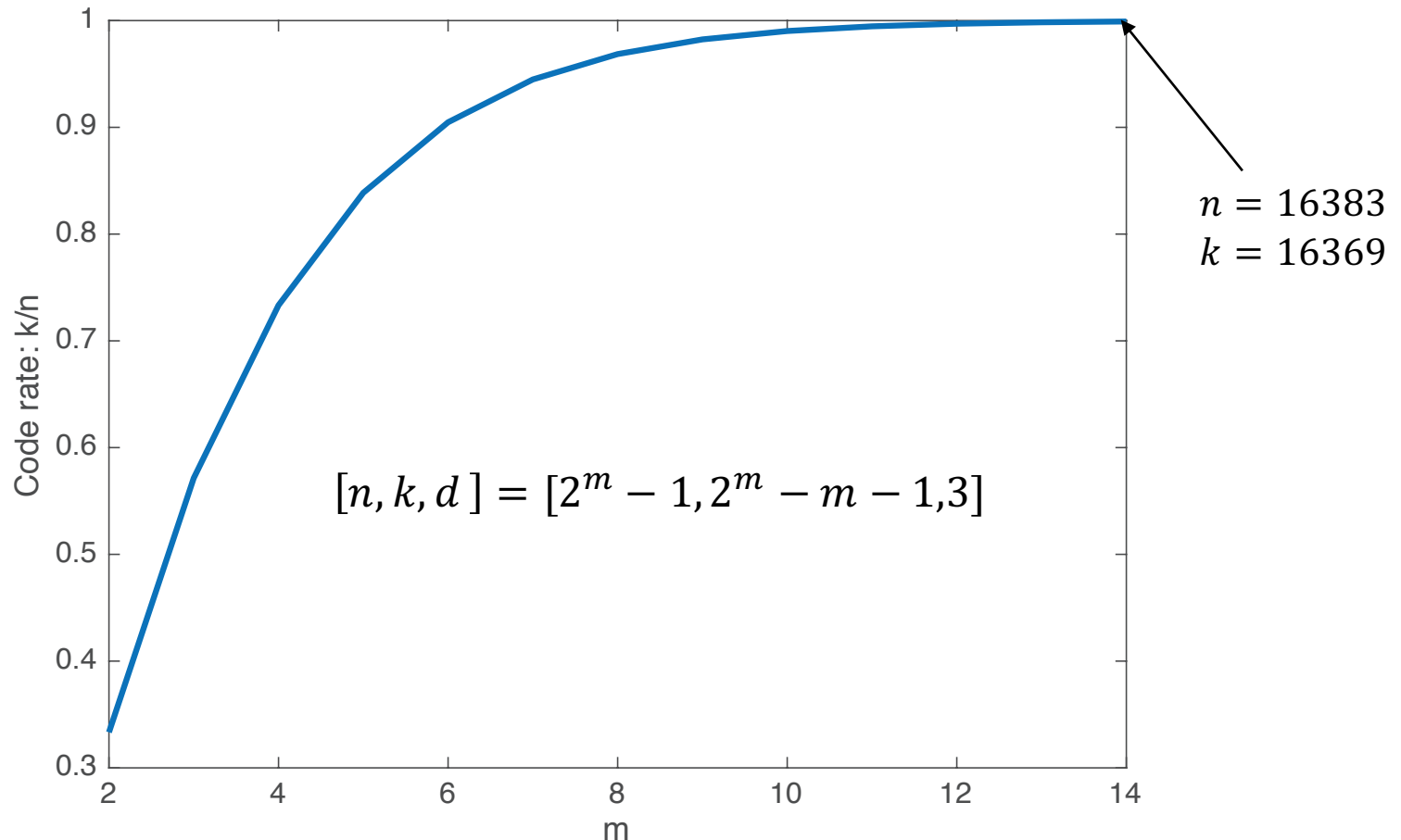
What is a Good Block Code? (1/2)

Consider: *A Hamming code*

- Defined by $m \geq 2$.
- Parity check matrix contains all non-zero binary vectors of length m .
- Gives: $n = 2^m - 1, \quad k = 2^m - m - 1$

$$[n, k, d] = [2^m - 1, 2^m - m - 1, 3]$$

What is a Good Block Code? (2/2)



Long block codes: Can achieve a given d with almost code rate 1

Binary Sequences

Consider: $a = a_0, a_1, a_2, \dots$ for $a_i \in \mathbb{F}_2$

- D -transform: Write sequence using delay operator D
- $A(D) = a_0 + a_1D + a_2D^2 + \dots$

Convolution of binary sequences:

- $P(D) = p_0 + p_1D + p_2D^2 + \dots + p_mD^m$
- Consider: $X(D) = A(D)P(D)$
- Time representation $x = a * p$:

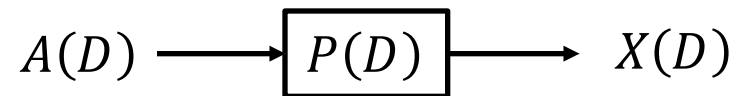
$$x_m = \sum_{k=0}^m a_k p_{m-k} = \sum_{k=0}^m a_{m-k} p_k$$

Filtering of Sequences

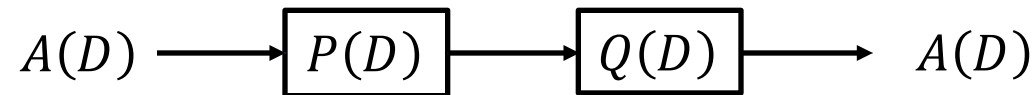
Input sequence: $A(D) = a_0 + a_1D + a_2D^2 + \dots$

Filter p :

- Impulse response: $p = p_0, \dots, p_m$
- Transfer function: $P(D) = p_0 + p_1D + p_2D^2 + \dots + p_mD^m$



Inverse filter q :



- Delay-less inverse q exist if and only if $p_0 = 1$
- Delayed inverses are also possible

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