

This is your last graded homework. Yippee! Write down your answers as clearly as possible, such that at least one of the assistants can decipher them. As usual, please write each problem on a separate piece of paper and don't forget to put your name on it as well. The due date for this homework is December 22nd 2009, at noon, INR-036.

PROBLEM 1. Assume that we use a code that maps the five numbers $\{1, 2, 3, 4, 5\}$ to the codewords $C = \{000000, 101100, 010011, 111111, 101010\}$.

- a) What is the rate of this code?
- b) Is this a linear code or not and why?
- c) What is the minimum distance of this code?
- d) How many erasures can it correct? How many errors can it correct? How many errors can it detect?

For parts (e-f) assume that we want to use this code to transmit over a noisy channel.

- e) In which of the following cases can the receiver correctly decode, and what does it decode?
 1. The receiver receives the sequence $(?, 1, ?, ?, 1, 1)$ (3 erasures)
 2. The receiver receives the sequence $(0, ?, ?, ?, ?, 1)$ (4 erasures)
- f) If the receiver receives the sequence $(1, 1, 1, 1, 0, 0)$, what is the minimum and what is the maximum number of errors that the channel has introduced?

PROBLEM 2. Let \mathcal{C} be an (n, k) linear code over a finite field F . Let H be a parity-check matrix of \mathcal{C} and v_1, v_2, \dots, v_n be the columns of H . Suppose that any set of d columns of H are linearly independent, where d is some positive integer number. Show that the minimum distance of \mathcal{C} is at least $d + 1$.

PROBLEM 3. Suppose that n, k, d are three positive integer numbers. Prove the following statements:

- a) If there exists an (n, k) binary linear code with minimum distance $2d + 2$, then there exists an $(n - 1, k)$ binary linear code with minimum distance $2d + 1$.
- b) If there exists an (n, k) binary linear code with minimum distance $2d + 1$, then there exists an $(n + 1, k)$ binary linear code with minimum distance $2d + 2$.

PROBLEM 4. In this problem you will use almost everything you learned in this class, signal processing, number theory, and last but not least channel coding.

In the first module of this course we have introduced the convolution of two signals as the fundamental operation which computes the effect of passing a signal through a filter.

Recall that if we have a signal $x[n]$ and the filter is given by $h[n]$, then the output of the filter is given by the convolution,

$$y[n] = \sum_m x[m]h[n - m].$$

Since convolutions are such basic operations, they are used very frequently. It is therefore important to find an efficient method to compute them. In this exercise we will learn the *Fourier transform* which is one of the most fundamental techniques in signal processing. It allows for an efficient implementation of convolutions and much more. The reason we have not introduced Fourier transforms in the signal processing module is that we need to be well familiar with complex numbers. But over finite fields it is much easier. We will also see that Reed-Solomon are most easily defined in terms of the Fourier transform. We will consider a very specific case but the same principle applies to any finite field.

- (a) Consider the field F_7 . Compute explicitly the set $\{5^i \pmod{7}\}, i = 0, 1, \dots, 5$. What is $5^6 \pmod{7}$? A field element a such that its power $a^i, i = 0, 1, \dots, |F| - 2$ covers all the non-zero elements of the field is called a *generator*.
- (b) Using your mastery of number theory (and the result in (a)), show that

$$\begin{aligned} 5^k &\equiv 1 \pmod{7}, \quad k = 0, \\ 5^k &\not\equiv 1 \pmod{7}, \quad 0 < k < 6, \\ 5^{6k} &= (5^6)^k \equiv 1 \pmod{7}, \quad k \in \mathbb{Z}. \end{aligned}$$

- (c) Show that for any integer k

$$(5^k - 1) \sum_{i=0}^5 5^{ki} = (5^k - 1)(1 + 5^k + 5^{2k} + 5^{3k} + 5^{4k} + 5^{5k}) = 5^{6k} - 1 = 0.$$

Argue that this implies that $\sum_{i=0}^5 5^{ki} \equiv 0 \pmod{7}$ for $0 < k < 6$ and that $\sum_{i=0}^5 5^{ki} \equiv 6 \pmod{7}$ for $k = 0$.

- (d) Consider a vector u of length 6 whose components are elements of F_7 . Define the following *Fourier transform* pair between the vectors u, \hat{u} of length 6 with components in F_7 . For $i, j = 0, \dots, 5$, let

$$\begin{aligned} \hat{u}_i &= \sum_{l=0,1,\dots,5} u_l 3^{il}, & \text{Fourier Transform} \\ u_j &= 6 \sum_{i=0,1,\dots,5} \hat{u}_i 5^{ij}, & \text{Inverse Fourier Transform} \end{aligned}$$

where all computations are done in F_7 . Note that 5 is the multiplicative inverse of 3.

- (e) Start with $u = (123456)$. Compute \hat{u} . Then verify that you get back u by computing the inverse transform.
- (f) Show that in general if you start with u and compute first \hat{u} by applying the Fourier transform and then compute the inverse Fourier transform of \hat{u} , you get back u .

To accomplish this, verify the following steps by using the results of (b) and (c):

$$\begin{aligned}
 u_j &= 6 \sum_{i=0,1,\dots,5} \hat{u}_i 5^{ij} \\
 &= 6 \sum_{i=0,1,\dots,5} \sum_{l=0,1,\dots,5} u_l 3^{il} 5^{ij} \\
 &= \sum_{l=0,1,\dots,5} u_l 6 \sum_{i=0,1,\dots,5} 5^{i(j-l)} \\
 &= \sum_{l=0,1,\dots,5} u_l 6 \sum_{i=0,1,\dots,5} (5^{j-l})^i \\
 &= u_j.
 \end{aligned}$$

- (g) Here is how you can use the Fourier transform to compute the convolution. Assume you have a signal $x[n]$ and a filter with impulse response $h[n]$. Assume you want to compute the *cyclic* convolution. This cyclic convolution differs slightly from the regular convolution in that we consider the signals to be periodic with period 6. In more detail, we have

$$y[n] = \sum_{m=0,\dots,5} x[m]h[n - m \bmod 6].$$

- (i) Compute the cyclic convolution of the two vectors $u = (123456)$ and $v = (121212)$.
- (ii) Compute the Fourier transform of u and v . Multiply the two resulting signals \hat{u} and \hat{v} componentwise, call the result \hat{w} . Take the inverse Fourier transform of \hat{w} . Hopefully, you get the same result as in (i). Indeed, this is the standard way a computer would compute the convolution.

In this example we considered a very simple example of a field with 6 non-zero elements. In practice you would use a much larger field, let's say with N non-zero elements. In this case the Fourier transform can be used to compute the cyclic convolution of periodic signals of period N . With a proper implementation of the Fourier transform (called the Fast Fourier transform) it is possible to compute the convolution in roughly $N \log N$ operations. A brute force implementation of the convolution on the other hand would require N^2 operations. This is a significant saving.

- (h) Now we will see how to define Reed-Solomon codes in terms of this Fourier transform. Define a RS code of length $n = 6$ with $k = 2$ and $d_{\min} = 5$ as follows. Consider the set of vectors \hat{c} of length 6 of the form $\hat{c} = (c_1, c_2, 0, 0, 0, 0)$, where $c_1, c_2 \in F_7$. The RS code is then the set of Fourier transforms of this set.
- (a) Show that this definition is equal to the definition we gave in class for the *canonical* Reed-Solomon code.
 - (b) Check that the following is a generator matrix for this code:

$$G = \begin{pmatrix} 1 & 1 & 1 & 1 & 1 & 1 \\ 1 & 3 & 2 & 6 & 4 & 5 \end{pmatrix}.$$

- (c) Assume that you want to transmit the information vector $u = (14)$. What is the corresponding codeword?
- (d) Assume you received the word $(?4?60?)$. You want to determine the received codeword. Write down the corresponding system of equations. Now use the Gaussian elimination algorithm to recover the transmitted information. What was the transmitted codeword?