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## Improved Decoding Scheme for Frequency-Hopped Multilevel FSK System

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We have recently examined, for possible application to digital mobile radio telephony, a digital spread-spectrum technique employing multiple frequency-shift keying (MFSK) modulation with codedivision-multiple access (CDMA) by frequency-hopping over a common bandwidth. The system uses the cellular approach where all mobiles within a cell communicate with a fixed base station in the cell. An analysis of base-to-mobile transmission shows that mutual interference limits the number of users which the system can accommodate at a given error rate. This paper describes a new decoding scheme to reduce mutual interference which makes use of the well-defined algebraic structure of the users' addresses. Analysis of the new decoder at high signal to noise (s/n) ratio shows it to outperform conventional decoding, allowing a 50 to 60 percent increase in the number of users who can simultaneously share the system at a given error rate. We describe a simple implementation of the decoder using shift registers.

#### I. INTRODUCTION

We have examined a digital spread-spectrum modulation technique employing frequency-hopping and multiple frequency-shift keying (MFSK) for multiple-access satellite communication<sup>1</sup> and for digital mobile radio telephony.<sup>2</sup> Every T seconds each user conveys a K-bit message by transmitting a sequence of L tones chosen from an alphabet of  $2^K$  sine waves of duration  $\tau (= T/L)$ . Code-division multiplexing is

used, and the message is modulated onto the address (code) assigned to each user.

The receiver, knowing the address, can decode the received signal and extract the message. However, transmissions by other users can combine to cause an erroneous message, resulting in an ambiguous reception. Thus even without channel impairments, the number of simultaneous users the system can support at a given error probability is interference limited.

The interference can be minimized by a proper choice of addresses with minimum cross correlation. Schemes for assigning  $2^K$  addresses which guarantee minimum mutual interference between  $2^K$  or fewer users have been proposed.<sup>3</sup>

The performance of the system could be improved, in principle, by decoding the addresses of all possible users and analyzing the interference pattern. Such a complete decoding scheme is very complex.

This paper describes a simpler decoder which makes use of the well-defined algebraic structure of the sequences<sup>3</sup> to eliminate erroneous messages which come from interference. The result is a significant improvement in the performance of the system with a much smaller complexity than complete decoding. If, for example, the system has a total one-way bandwidth of 20 MHz and the data rate of each user is 32 kilobit/second, the number of users which the system can support at a bit error probability of  $10^{-3}$  is increased by 60 percent over that attainable with conventional decoding.

In the performance analysis of the new decoder, an ideal (error-free) channel has been assumed. The conventional scheme has been further analyzed<sup>2</sup> for base-to-mobile transmission through noisy and multipath fading channels and was found to degrade gracefully. Under the same conditions, the new decoder is expected to have a similar degradation attaining its substantial improvements over the conventional decoder.

#### II. SYSTEM DESCRIPTION

The elementary signals of the system are a set of  $2^K$  sine waves which are orthogonal over the chip duration  $\tau$ . Every T seconds each user conveys a K-bit message by transmitting a sequence of L tones of duration  $\tau$  chosen from the signal set. The sequence is uniquely determined by the user's address and the K-bit message and can be described as a pattern in the  $L \times 2^K$  time-frequency matrix A. Simultaneous transmissions by M users can result in up to  $L \times M$  entries in A (Fig. 1).

Let the address of the mth user be denoted by a vector  $\mathbf{a}_m$ :

$$\mathbf{a}_m=(a_{m1}, a_{m2}, \cdots, a_{mL}),$$

where each  $a_{mi}$  is a K-bit number corresponding to one of the  $2^{K}$  frequencies of the system.

The transmitted sequence  $Y_m$  is formed by modulating the message  $X_m$  onto  $a_m$ . The message can modulate the address in different ways. In Refs. 1 and 2, this is done by summing (mod  $2^K$ ) the address and the message to obtain

$$\mathbf{Y}_m = \mathbf{a}_m + X_m \cdot \mathbf{1},\tag{1}$$

where

$$\mathbf{1} = (\underbrace{1, 1, \cdots, 1}_{L}).$$

Other schemes are possible (see, for example, Ref. 3) as long as  $\mathbf{Y}_m$  can be uniquely decoded.

Every  $\tau$  seconds the receiver performs a spectral analysis of the composite received signal Y and decides which of the  $2^K$  frequency cells contain energy. Thus after T seconds, assuming no channel impairments, a duplicate of A is generated at the receiver.

To decode  $X_m$  the receiver performs the inverse operation of (1), i.e., cycle-shift each column i of A by  $-a_{mi}$  to obtain user m's decoded matrix  $A_m$  (Fig. 1). The message  $X_m$  will appear as a complete row in  $A_m$ .

Ambiguous decoding occurs when transmissions by other users combine to form other complete rows in  $A_m$ . In a previous analysis of the system, <sup>1,2</sup> it was assumed that if there is more than one complete row the receiver has no way of finding the correct message and has to

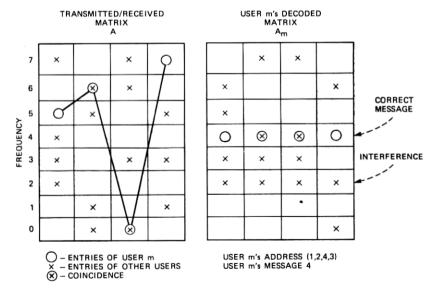


Fig. 1—Transmitted/received matrix A and user m's decoded matrix  $A_m$  for K=3 and L=4.

choose one of them at random. We show how further investigation of the complete rows can improve the performance.

## III. ADDRESS ASSIGNMENT

It is convenient to describe the modulation/decoding procedure in algebraic terms by denoting the  $2^K$  frequencies as elements of  $GF(2^K)$ , which is the finite field (Galois field) of  $2^K$  elements 0, 1,  $\cdots$ ,  $2^K - 1$ . Such finite fields exist for all  $Q = p^n$ , where p is a prime and  $n = 1, 2, \cdots$ , and the following analysis and results can be extended to systems with Q frequencies.

Accordingly, the K-bit message  $X_m$ , the components of  $\mathbf{a}_m$ ,  $Y_m$ , and the row numbers in A,  $A_m$  can all be expressed as elements of  $GF(2^K)$ , and the operations of addition and multiplication are performed according to the rules of  $GF(2^K)$  (see the appendix).

Reference 3 describes an optimum structure of addresses which guarantees minimum mutual interference between  $2^K$  or fewer users. In this scheme, each user is assigned a distinct element of  $GF(2^K)$ . Let  $\gamma_m$  denote the element assigned to user m and let  $\beta$  be a fixed primitive element in  $GF(2^K)$ . The address of user m is defined to be

$$\mathbf{a}_m = (\gamma_m, \gamma_m \beta, \gamma_m \beta^2, \cdots, \gamma_m \beta^{L-1}). \tag{2}$$

Since there are  $2^K$  distinct elements  $\gamma_m$ , the maximum number of users (distinct addresses) is  $2^K$ . Assuming word synchronization (i.e., all users start their sequences at the same time slot) it has been shown that for  $L \leq 2^K - 1$  the transmitted sequences of any two users will coincide in at most one chip for all possible message values.

As mentioned before, other modulation schemes are possible (e.g.,  $\mathbf{Y}_m = \{y_{mi}\}, y_{mi} = X_m \beta^{i-1} + \gamma_m$ ), and the analysis and results that follow can be extended to such systems.

#### IV. NEW DECODING SCHEME

## 4.1 Decoding procedure

Let M be the number of simultaneous users of the system. Each user i transmits a sequence

$$\mathbf{Y}_i = \mathbf{a}_i + X_i \cdot \mathbf{1},$$

where the address  $\mathbf{a}_i$  is assigned according to (2) and the message value is  $X_i$ . We assume a synchronous transmission, i.e., all users start their sequences at the same time slot.

In the decoded matrix  $A_m$  of user m,  $X_m$  appears as a complete row  $X_m \cdot 1$ . Suppose we have another complete row X' in  $A_m$  which is the result of interference. According to the address assignment, each chip in X' must come from a different user, so at least L users  $i_1, \dots, i_L$  have combined to cause this interference. The user that contributes

the interference in column n of X' is denoted by  $i_n$ . (If more than one user caused this interference,  $i_n$  can be any one of them.)

From eqs. (1) and (2), the *j*th entry  $(j = 1, \dots, L)$  of user  $i_n$  in  $A_m$  will be at row  $c_n(j)$  and column j where

$$C_n(j) = \gamma_{i_n} \beta^{j-1} + X_{i_n} - \gamma_m \beta^{j-1},$$
  
=  $\delta_n \beta^{j-1} + X_{i_n}, \quad j = 1, \dots, L,$  (3)

 $\gamma_m$  is the address of the mth user and

$$\delta_n \triangleq \gamma_{i_n} - \gamma_m. \tag{4}$$

Note that  $\delta_n \neq 0$  for all interferers  $i_n$ ,  $n = 1, \dots, L$ .

 $C_n(n)$  is the row in which the entry of user  $i_n$  appears at the *n*th column. By definition, this row is X'. So we have the following relation:

$$C_n(n) = \delta_n \beta^{n-1} + X_{i_n} = X'.$$
 (5)

Let  $D_X^*$  be the matrix obtained by subtracting the number X' from all rows of  $A_m$  (equivalently, shifting row X' in  $A_m$  to row zero). At  $D_X^*$  the entry of user  $i_n$  at column j  $(j = 1, \dots, L)$  will be at row  $d_n(j)$ . From (3) to (5) we have

$$d_n(j) = c_n(j) - X'$$
  
=  $\delta_n(\beta^{j-1} - \beta^{n-1}), \quad n, j = 1, \dots, L.$  (6)

Thus by shifting row X' to row zero, the locations of the entries of the interferers  $i_1, \dots, i_L$  are independent of their message values  $X_{i_1}, \dots, X_{i_L}$ .

So if row X' in  $A_m$  was formed by L interferers with addresses  $\delta_n + \gamma_m$ , their entries will appear in  $D_{X'}^*$  at rows  $d_n(j)$  and column j  $(j, n = 1, \dots, L)$ . We thus have the following theorem:

Theorem: A necessary condition for row X' in  $A_m$  to be caused by interference is the existence of L nonzero numbers  $\delta_n$  in  $GF(2^K)$  such that all entries  $(d_n(j), j)(n, j = 1, \dots, L)$  appear in  $D_X^*$ .

Using (6) we can write for  $n = 1, \dots, L - 1$ :

$$d_n(j) = \delta_n \beta^{n-1} (\beta^{j-n} - 1),$$

$$d_n(n+1) = \delta_n \beta^{n-1} (\beta - 1).$$
(7)

Thus we have the following relation:

$$d_n(n+1) = f_{i-n}d_n(j), (8)$$

where

$$f_{j-n} \triangleq \frac{\beta-1}{\beta^{j-n}-1}, \quad j=1,\ldots,L, \quad j\neq n,$$
 (9)

and all the  $f_{j-n}$  are independent of the particular user and are fixed for the system. To check for the existence of user  $i_n$   $(n=1,\dots,L-1)$ , we multiply all entries (i.e., their row numbers) in column j  $(j=1,\dots,L,j\neq n,j\neq n+1)$  of  $D_X^*$  by  $f_{j-n}$  and look for a "complete" row (complete except for the nth term). If we find such a row, then a possible interferer contributing to column n of X' exists. We see in Section 4.2 that there is a simple way to realize this procedure.

To check for user  $i_L$ , we modify this procedure and use the formula

$$d_L(L-1) = f_{j-L}^* d_L(j), \qquad j = 1, \dots, L-2,$$
 (10)

where

$$f_{j-L}^* \triangleq \frac{1}{\beta} f_{j-L}. \tag{11}$$

We then multiply all entries in column j ( $j = 1, \dots, L-2$ ) by  $f_{j-L}^*$  and check for a complete row.

If the condition is satisfied for all  $n = 1, \dots, L$ , we assume that row X' in  $A_m$  is the result of interference.

Using this decoding scheme, all interference rows X' will be identified as such. The correct row  $X_m$  will usually fail to satisfy the condition for some n and thus will be identified and decoded as the correct message.

## 4.2 Realization

Multiplication of two elements in  $GF(2^K)$  is most easily performed by expressing each term as a power of a primitive element  $\beta$  in the field (see the appendix).

Let  $D_{X'}$  denote the matrix  $D_{X'}^*$  without row zero. Each row q in  $D_{X'}$  can be expressed by its exponent q', that is,

$$q = \beta^{q'}, \qquad q, q' = 1, \dots, 2^K - 1.$$
 (12)

Since  $\beta$  is primitive in  $GF(2^K)$ , the transformation  $q \to q'$  is a row permutation  $D_{X'} \to P_{X'}$ .

Similarly, we can express the multiplication factors  $f_i$ ,  $f_i^*$  as powers of  $\beta$ :

$$f_i = \beta^{k_i}, \qquad f_i^* = \beta^{k_i - 1}.$$
 (13)

Multiplying an element at row q and column j in  $D_{X'}$  by  $f_{j-n}$  is equivalent to cycle-shifting the corresponding element in  $P_{X'}$  (which appears at row q' and column j)  $k_{j-n}$  positions.

So to find the existence of a user  $i_n$  contributing to column n of X', we cycle-shift all entries in column j of  $P_{X'}$  (for  $j = 1, \dots, L, j \neq n$ ,  $j \neq n + 1$ )  $k_{j-n}$  positions and look for a complete row.

Note that the rule of permutation  $q \to q'$  as well as the shifting operator  $k_i$ ,  $i = \pm 1, \pm 2, \dots, \pm (L-1)$  are fixed for the system and

are independent of the interference pattern. Thus they can be computed once and stored at the receiver.

In comparing the complexity of this decoding scheme to complete decoding, we have here L + L(L-2) cycle-shifting operations for each complete row of  $A_m$ , while in complete decoding we perform  $L(2^K-1)$ cycle-shifting operations. A similar relation holds for the number of checking complete rows. So the complexity of the proposed scheme is of the order of  $L^2$  compared to an order of  $L2^{K}$  for the complete decoding. Since  $L \approx K \ll 2^K$ , the new decoding scheme has a much smaller complexity than complete decoding, yet it achieves a substantial improvement in performance.

## V. DECODING ALGORITHM OF USER m

## 5.1 Preliminary computations

These computations are independent of the received matrix and are identical for all users so they can be performed once and stored at the

(i) Express all nonzero elements  $q \in GF(2^K)$  by their exponents q', i.e.,  $q = \beta^{q'}$ , where  $\beta$  is the primitive element in  $GF(2^{\bar{k}})$  used to generate the addresses. This results in a table of  $2^K - 1$  pairs (q, q').

(ii) Compute 2(L-1) terms  $k_i$ ,  $i=\pm 1,\pm 2,\cdots,\pm (L-1)$ , where

$$\beta^{k_i} = \frac{\beta - 1}{\beta^i - 1},$$

and generate the cycle-shifting matrix  $S = \{S_{i,i}\}$ ,

$$S_{i,j} = \begin{cases} 0 & i = 1, \dots, L & j = i \\ k_{j-i} & i = 1, \dots, L - 1 & j = 1, \dots, L \\ & & j \neq i \\ k_{j-L} - 1 & i = L & j = 1, \dots, L - 1. \end{cases}$$
(14)

## 5.2 Computation performed for each code word (L chips)

- (i) From the received matrix A, compute the decoded matrix  $A_m$  by cycle-shifting the columns of A according to the address  $a_m$ . If  $A_m$ contains only one complete row at  $X_m$ , it is decoded as the message and the procedure terminates. If there are I complete rows ( $I \ge 2$ ),  $X^{(1)}, \dots, X^{(I)}$  we check each row  $X^{(i)}$  as follows. (ii) Subtract (the number of row)  $X^{(i)}$  from all rows in  $A_m$  and discard
- row zero to obtain matrix  $D_X$ .
- (iii) Perform row permutation of  $D_X \rightarrow P_X$  according to the table of  $2^K - 1$  pairs (q, q').
- (iv) For each n  $(n = 1, \dots, L)$ , cycle-shift all columns j in  $P_X$  $(j = 1, \dots, L, j \neq n, n + 1)$   $S_{n,j}$  positions, according to the cycle-

shifting matrix S [eq. (14)] and look for a complete row (except the term in the nth column).

If for some n we don't find such a complete row, we decode

$$X_m = X^{(i)},$$

and the procedure terminates. Otherwise we assume  $X^{(i)}$  to be the result of interference and repeat steps (ii) to (iv) for  $X^{(i+1)}$ .

(v) If all rows  $X^{(i)}(i=1,\dots,I)$  pass the interference test (this could happen, for example, if an interference row coincides with the correct message), we cannot determine the correct message and have to pick one of the  $X^{(i)}$  at random, as is always done in conventional decoding.

A flow diagram of the decoding scheme is outlined in Fig. 2.

The average time needed to decode  $X_m$  can be reduced by performing the interference test in a different order. Instead of checking the complete rows sequentially (as described above), we can check the first chip of all complete rows, then the second chip, and so on, until we find a chip that does not pass the interference test. It can be shown that with this scheme the average number of chips checked will be smaller and thus the average decoding time will be reduced.

### VI. DECODER IMPLEMENTATION

The decoder can be implemented either in software or in hardware. A software implementation is appropriate for a microprocessor-controlled receiver and is done according to the flow diagram of Fig. 2.

Since most decoding operations require shifting [cycle-shifting or subtraction in  $GF(2^K)$ ] of columns of various matrices, a simple way to implement the decoder in hardware is by using shift registers.

The decoder is composed of L identical units  $U_1, \cdots, U_L$  representing the L columns of the matrices and a single L-stage shift register used to count the number of entries in a row and find the complete rows (Fig. 3a). A possible implementation of the basic unit  $U_j$  is depicted in Fig. 3b. The left shift register system  $R_{j1}$  contains K+2 registers of length 1, 1, 1, 1, 4, 8,  $\cdots$ ,  $2^{K-1}$  (for a total of  $2^K$  cells) and K identical double-pole double-throw switches. When all switches are in position 1, cycle-shifting can be performed. To subtract [according to the rules of  $GF(2^K)$ ] a number  $2^{i-1}$ , each cell in (binary representation) position  $(\delta_K, \cdots, \delta_i, \cdots, \delta_i, \cdots, \delta_1)$  has to be shifted to position  $(\delta_K, \cdots, \delta_i, \cdots, \delta_1)$ . For  $1 \le i \le K-1$ , this is accomplished by shifting the register  $2^K$  steps while changing the position of the switch number i every  $2^{i-1}$  steps (for i=1, 2 we start at position 1 and for  $3 \le i \le K-1$  at position 2). For i=k, the register is cycle-shifted  $2^{K-1}$  steps.

In general, to subtract a number  $N = (n_K, \dots, n_1)$ , all the switches corresponding to the nonzero coefficients of its binary representation (except  $n_K$ ) are operated while the register is shifted  $2^K$  steps. (Switch

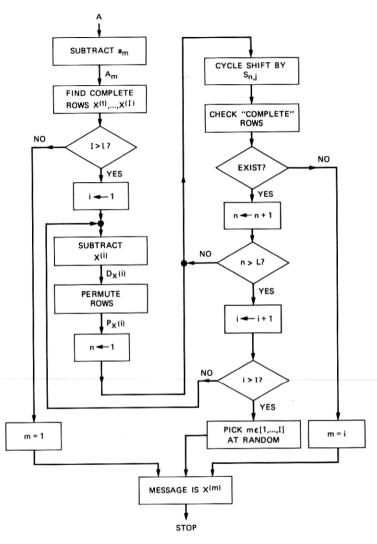
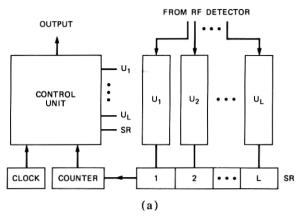


Fig. 2—Flow diagram of the new decoder. The input is the received matrix A. The output is a decoded message value  $X_m$ .

1b is used instead of 1a whenever switch 2 moves to position 2.) If  $n_K = 1$ , the register is cycle-shifted additional  $2^{K-1}$  steps. An example for N = 3 and N = 7 is shown in Table I.

The right shift register  $R_{j2}$  contains  $2^K - 1$  cells and performs cycle shift operation. The data are fed from the left register system to the right register in a parallel-to-parallel mode. Each stage in  $R_{j2}$  is hardwired to a stage in  $R_{j1}$  according to the row permutation (fixed) rule  $q \rightarrow q'$ .



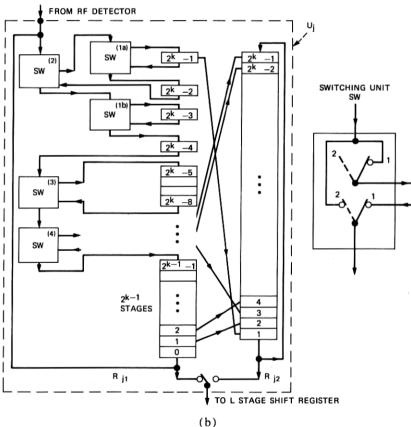


Fig. 3—Hardware implementation of the decoder. (a) L identical units  $U_j$  operate in parallel. SR is used to find the complete rows. (b) The basic unit  $U_j$  includes two shift register systems:  $R_{j1}$  can perform subtraction in  $GF(2^K)$  using the switching units sw, and  $R_{j2}$  perform additional decoding, when necessary, to remove ambiguity.

Table I—K = 3 (8 cells, 3 switches)

				•		•							
	0	1	2	3	4	5	6	7	8	9	10	11	12
Switch													
1a		1	2	1	1	1	<b>2</b>	1	1	1	1	1	1
1b		1	1	1	1 2 2	1	1	1	2	1	1	1	1
2		1	1	2	2	1	1	2	2	1	1	1	1
Content of Register													
7	7	0	0	0	0	4	4	4	4	3	2	1	0
6	6	7	1	1	1	0	5	5	5	4	3	2	1
5	5	6	7	<b>2</b>	2 3	1	0	6	6	5 6	4	3	2 3
4	4	5	6	7	3	2	1	0	7	6	4 5 6	4	3
3	3	4	5	6	7	3	2 3	1	0	7	6	5	4
2	2	3	4	5	6	7	3	2	1	0	7	6	5
1	1	2	3	4	5	6	7	3	2	1	0	7	6
0	0	1	2	3	4	5	6	7	3	2	1	0	7
								Cycle					
										l	shi	ifting	3
Sı	ıbtrac	t 3 =	: (1,	1) —					<b>—</b> →				
	Sub	trac	t 7 =	(1,	1, 1)								

To decode a block, the data are serially fed from the RF detectors to the left register systems  $R_{j1}$  ( $j=1,\dots,L$ ), which are then shifted (with switch positions corresponding to the address  $a_{mj}$ ) to generate  $A_m$  and find its complete rows. To check a complete row, it is shifted to row zero and the data are transferred to the right registers  $R_{j2}$ , which are then cycle-shifted according to the shifting matrix S.

The L-stage shift register SR is parallel-fed from either  $R_{j1}$  or  $R_{j2}$  and is used to count the number of entries in a row to find the complete rows.

#### VII. ERROR PROBABILITIES

Let the system parameters be K ( $2^K$  frequencies), L (sequence length), and M (simultaneous users). We assume an algebraic address assignment [such as (2)] and a synchronous transmission without channel impairments.

If the decoded matrix  $A_m$  of user m has only one complete row, this row is correctly decoded as the message value  $X_m$ . If there are two or more complete rows in  $A_m$ , an interference test is performed. An interference row will always pass the test since each of its chips is part of a sequence transmitted by some user. The correct row will usually fail to pass the test for some of its chips and thus can be identified and correctly decoded.

We will have an ambiguity in decoding when the following two conditions are satisfied:

- (i) There are two or more complete rows in  $A_m$ .
- (ii) The correct row passes the interference test.

When this happens, we choose one complete row at random and decode it as the message.

Let  $P_1$  and  $P_2$  be the probabilities of condition (i) and (ii), respectively. Although not strictly independent, it can be shown using random-coding arguments that the two conditions can be assumed to be independent with a negligible effect on the probability of error.  $P_1$  can be upper bounded by the union bound

$$P_1 < (2^K - 1)p^L, (15)$$

where

$$p = 1 - (1 - 2^{-K})^{M-1}. (16)$$

Condition (ii) can happen as follows: j chips in  $X_m$  ( $j = 0, 1, \ldots, L$ ) are also the result of interference (by at least j other users) and the remaining L - j chips, although not caused by interference, pass the interference test. That is, entries from other users combine to form complete rows in the cycle-shifted version of  $P_{X_m}$ .

Let  $P_{2,j}$  be the probability of condition (ii) when exactly j chips in  $X_m$  come from interference, then

$$P_2 = \sum_{j=0}^{L} P_{2,j}.$$
 (17)

 $P_{2,L}$  corresponds to the case where interference from other users combine to form a complete row which coincides with  $X_m$ . Thus

$$P_{2,L} < p^L. \tag{18}$$

Similarly,

$$P_{2,L-1} < Lp^{L-1}(1-p)\hat{p}, \tag{19}$$

where  $Lp^{L-1}(1-p)$  is the probability that exactly L-1 chips come from interference and  $\hat{p}$  is the probability that a complete row (L-1 entries) exists in the cycle-shifted  $P_{X_m}$ , when checked for the remaining chip. Since this complete row could be any of the  $2^K-1$  rows of  $P_{X_m}$ , we have

$$\hat{p} < (2^K - 1)p^{L-1}. (20)$$

The general term  $P_{2,L-j}$  can be upper bounded by

$$P_{2,L-j} < p^{L-j} (1-p)^{j} {L \choose i} \hat{p}^{j}.$$
 (21)

Thus

$$P_{2} < p^{L} \sum_{j=0}^{L} {L \choose j} \left[ (2^{K} - 1)(1 - p)p^{(L-2)} \right]^{j},$$

$$P_{2} < p^{L}(1 + S)^{L},$$
(22)

where

$$S = (2^{K} - 1)(1 - p)p^{(L-2)}.$$
 (23)

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So, as long as p(1 + S) < 1, we can upper-bound the word error probability  $P_E$  by

$$P_E < \frac{1}{2} P_1 \cdot P_2 < \frac{1}{2} (2^K - 1)(1 + S)^L p^{2L},$$
 (24)

where the factor of ½ comes from the random choice of message when the decoding fails. From this we can obtain a bound on the bit error probability<sup>1</sup>

$$p_b < 2^{K-2}(1+S)^L p^{2L}. (25)$$

Comparing the performances to conventional decoding, we see that the new decoding scheme has nearly doubled the exponent of p.

## VIII. SYSTEM PERFORMANCE

The bandwidth necessary to support  $2^K$  tones that are orthogonal over  $\pi = T/L$  seconds is

$$W = 2^K/\tau = 2^K L/T \text{ Hz.}$$

The transmission rate per user is

$$R = K/T$$
 bit/s.

For a given W, R, and K, the length of the sequence L is determined by

$$L=rK2^{-K},$$

where

$$r = W/R$$
.

Let M be the number of users that the system can accommodate at a given error probability  $P_b$ . The efficiency  $\eta$  of the system is defined as

$$\eta = MR/W = (M/r)$$
 bit/s/Hz,

the information rate (bit/s) per unit bandwidth transmitted through the system with a bit error probability  $P_b$ . For a given W, R, and  $P_b$ , there is an optimum K which maximizes the efficiency.

Figure 4 shows the dependence of M (and  $\eta$ ) on K for  $P_b = 10^{-3}$  and r = 626, 320, 165 (corresponding to R = 32 kilobit/second and  $W \approx 20$ , 10, and 5 MHz, respectively) using the upper bound (25) on  $P_b$ .

Since the decoding scheme is based on the address assignment (3) which can accommodate at most  $2^K - 1$  users, M cannot exceed this number. This is shown by the dashed curves. As can be seen, this has no influence on the choice of optimum K and the performance at that K.

Figure 5 shows the efficiency of the system as a function of r for 300  $\leq r \leq$  700, when the optimum K was taken at each r. It can be seen that, since K, L, and M assume only integer values, the curve is not

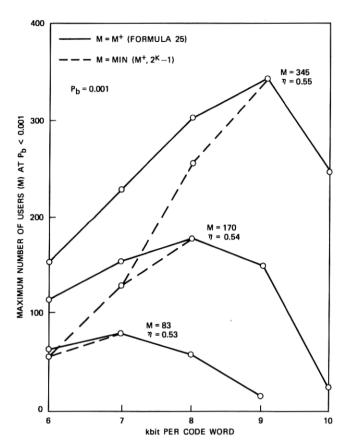


Fig. 4—Number of users M which can simultaneously share the system at an error probability of 0.001 as a function of K, for different values of r = W/R. The solid line  $(M^+)$  is computed using the upper bound (25). The broken line indicates the limit on M ( $2^K$ ) when the address assignment (2) is used.

continuous but has local maxima at specific values of r. The same phenomenon occurs for conventional decoding.

To compare the performance of the new decoding scheme with that of conventional decoding, consider the case where  $W=20.025~\mathrm{MHz}$ ,  $R=32~\mathrm{kilobit/second}$  (r=625.8), and  $P_b=10^{-3}$ . (At this r, the efficiencies of both schemes have local maxima.) The optimum K is 9 (corresponding to 512 frequencies) and L=11 in both schemes. The maximum number of users which the system can accommodate is 345 in the new scheme compared to 216 with conventional decoding. The corresponding efficiencies are 55 and 34.5 percent, an improvement of almost 60 percent. A similar improvement (of 50 to 60 percent) is obtained throughout the range of r.

Figure 6 depicts the bit error probability as a function of the number

of users for both decoding schemes for the above parameters (K = 9, L = 11, and r = 625.8).

While the number of users which the system can actually support would be less than illustrated here because of the idealized model used in the analysis (as discussed in Ref. 2), the comparative results are believed to be valid.

## XI. SUMMARY AND CONCLUSIONS

A new decoding scheme which makes use of the algebraic structure of the addresses to eliminate erroneous messages which come from interference was described. Analysis of the noiseless case, where interference from other users is the only source of performance impairments, shows a 50 to 60 percent improvement in efficiency over conventional decoding.

In noisy and multipath fading channels, the received matrix is corrupted by insertions and deletions of entries and the correct row might be incomplete. In this case, the row with the maximum number of entries is decoded as the correct message, and the probability of ambiguous reception is increased. For base-to-mobile transmission, the conventional decoder has been analyzed<sup>2</sup> in Gaussian noise and a multipath fading environment and found to degrade gracefully. Other

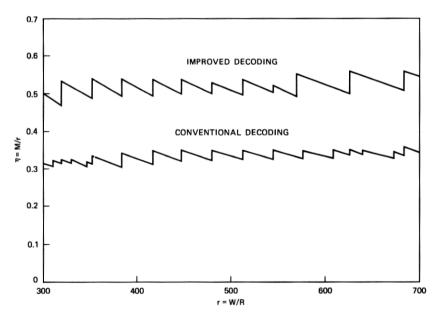


Fig. 5—The efficiencies  $(\eta = M/r)$  for both decoding schemes as a function of r = W/R, when the optimum K is taken at each r. Since K, L, and M assume only integer values the curve is not continuous.

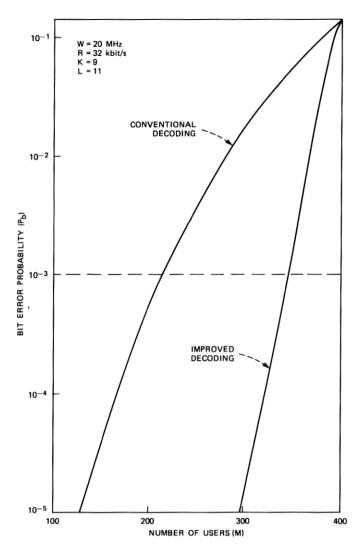


Fig. 6—Performance curve as a function of the number of users M for conventional and improved decoding. The system parameters are W=20 MHz, R=32 kilobit/second, K=9, and L=11.

factors such as shadow fading, intercell interference (when the system is embedded in a matrix of cells covering a large geographic area), and synchronization error due to multipath propagation delay spread (most relevant to mobile-to-base transmission) are likely to further reduce the number of users but as yet have not been analyzed. The new decoder is likely to have a similar degradation in typical noisy conditions encountered in mobile radio and satellite communication sys-

tems, but can be expected to yield a substantial improvement over conventional decoding.

#### **APPENDIX**

## Algebra of Finite Fields

## A.1 Rules of addition (subtraction) and multiplication (division)

Let GF(Q) be a finite field (Galois field) of Q elements  $0, 1, \dots,$ (Q-1). Such fields exist for all  $Q=p^n$ , where p is a prime and n=1, 2, ....

If Q is a prime (n = 1), the rules of addition and multiplication are defined by ordinary modulo Q arithmetics.

If  $Q = p^n$ ,  $n = 2, 3, \cdots$  we first represent the elements as p-ary vectors of length n (if p = 2, this is the binary representation). Addition is now defined as mod p addition of the components. To specify multiplication, the n-tuplets are transformed into polynomials of degree n-1 in Z by letting the first digit be the coefficient of  $Z^{n-1}$ , and so on. The multiplication rule is polynomial multiplication (ordinary multiplication with mod p addition of coefficients) modulo an irreducible polynomial of degree n.

## A.2 Multiplication using an exponent representation

A simple way to perform multiplication in  $GF(p^n)$  is: Every nonzero element in the field can be expressed as a power of a primitive element  $\beta(\beta)$  is primitive if the smallest integer k such that  $\beta^k = 1$  is k = Q - 1. Thus if we have a list (table) of all exponent representations, we can perform a multiplication of two numbers by adding, modulo (Q-1), their exponents. This is analogous to multiplication of real numbers using a table of logarithms.

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