

Underwater Acoustics Signal Processing Digital Signal Processing Design of Computationally Efficient Elliptic IIR Filters with a Reduced Number of Shift-and-Add Operations New Precoding for Intersymbol Interference Cancellation Using Nonmaximally Decimated Multirate Filterbanks wi Magnitude Response Peak Detection and Control Using Balanced Model Reduction and Leakage to a Targ On Properties of Information Matrices of Delta-Operator Based Adaptive Signal Processing Algorithms..... Statistical Signal and Array Processing Joint Estimation of Time Delays and Directions of Arrival of Multiple Reflections of a Known Signal Efficient Method for Estimating Directions-of-Arrival of Partially Polarized Signals with Electromagnetic Vector Multidimensional Signal Processing **Emerging Techniques** (Contents Continued on Bac.

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precoding methods, such as the TH and trellis precodings, the new precoding

- i) may be independent of the ISI channel;
- ii) is linear and does not have to implement any modulo operation;
- iii) gives the ideal FIR equalization at the receiver for any FIR ISI channel including spectral-null channels;
- iv) expands the transmission bandwidth in a minimum amount.

The precoding is built on nonmaximally decimated multirate filterbanks. Based on multirate filterbank theory, we present a necessary and sufficient condition on an FIR ISI transfer function in terms of its zero set such that there is a linear FIR $N \times K$ precoder so that an ideal FIR equalizer exists, where the integers K and N are arbitrarily fixed. The condition is easy to check. As a consequence of the condition, for any given FIR ISI transfer function (not identically 0), there always exist such linear FIR precoders. Moreover, for almost all given FIR ISI transfer functions, there exist linear FIR precoders with size $N \times (N-1)$, i.e., the bandwidth is expanded by 1/N. In addition to the conditions on the ISI transfer functions, a method for the design of the linear FIR precoders and the ideal FIR equalizers is also given. Numerical examples are presented to illustrate the theory.

I. INTRODUCTION

INTERSYMBOL interference (ISI) is a common problem in telecommunication systems, such as terrestrial television broadcasting, digital data communication systems, and cellular mobile communication systems. The main reasons for the ISI are because of high-speed transmission and multipath fading. There have been considerable studies for these problems, such as [1]–[29] and [33]–[40]. These studies can be primarily split into three categories:

- i) post equalization, such as least-mean-squared (LMS) equalizer and decision feedback equalization (DFE), for example, [1]–[3], [18]–[29], and [36]–[39];
- ii) multicarrier modulation to increase transmission symbol length, for example, [4]–[6];

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and Forney [9], [10], matched spectral null proin partial response channels [12], and other proschemes, for example, [13]–[17] and [40].

The basic idea for DFE is that once an information has been detected, the ISI that it causes on future symbol be estimated and subtracted out prior to symbol detectic usually consists of a feedforward filter and feedbacl The feedback filter is driven by decisions of the outpu detector, and its coefficients are adjusted to cancel the the current symbol that results from past detected symbol coefficient adjustment may be done via a linear equaliz LMS algorithms. The convergence of these iterative algorithms are dependent of the channel characteristics. When a cha spectral null or frequency selective fading, these algorith very slow and, therefore, become computationally exp The performance of the existing linear equalizers signi degrades over frequency selective fading channels. A DFE has better performance than the existing linear equ when the frequency fading is in the middle of a pa it does not offer much improvement in other fading For more details, see, for example, [3] and [35]. equalization techniques, there are many research resul for example, [18]-[29] and [36]-[39] on blind equality where channel characteristics are assumed unknown. I equalization techniques, there are approximately three of results:

- i) high-order statistics techniques;
- second-order cyclostationary statistics technique oversampling;

iii) antenna array (smart antenna) multireceiver tech where there is a considerable amount of overlaps betv and iii).

A block diagram for TH precoding is shown in Fig. 1 the basic idea is to equalize the signal before transm With TH precoding there are two drawbacks: i) The tranneeds to know the channel characteristics, and ii) the preis not reliable when the ISI channel H(z) has spectral frequency selective fading characteristics, which is b the pre-equalizer mod[1/H(z), M] oscillates in a dr way when H(z) is close to zero. The trellis precoding a proposed by Eyuboglu and Forney [9] whitens the noise equalizer output. This scheme combines precoding and

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Fig. 2. Nonmaximally decimated multirate filterbank in a communication channel with ISI.

shaping. There are also similar drawbacks about this approach.

- i) The transmitter also needs to know the ISI channel characteristics.
- ii) The trellis shaping depends on the ISI channel.
- iii) The trellis precoding technique may not be suitable for spectral-null channels either.

In the matched spectral null precoding scheme [12] in partial response channels, certain error control codes are chosen to match the spectral nulls of partial response channels in order to lose less signal information through the channel. This approach is mainly for magnetic recording systems.

We now propose a multirate filterbank as a precoder before transmission (shown in Fig. 2), where $\downarrow K$ indicates downsampling by K, and $\uparrow N$ indicates upsampling by N, i.e., inserting N-1 zeros between two adjacent samples, and H(z) is the ISI transfer function. Later, we will see a multirate filterbank decoder for the receiver to eliminate the ISI. If input signal x[n] in Fig. 2 can be completely recovered from the received signal $\hat{x}[n]$ through an FIR linear system, we call that the system in Fig. 2 has perfect reconstruction (PR) or an FIR ideal linear equalizer. In what follows, we use "precoder" and "multirate filterbank" interchangeably.

With the precoder proposed in Fig. 2, there are three questions to be answered:

- i) What is the condition on H(z) such that there exists a multirate filterbank with N channels and decimation by K in Fig. 2 so that x[n] can be recovered from $\hat{x}[n]$ through an FIR linear system?
- ii) If the condition on H(z) in the first question is satisfied, how does one design a multirate filterbank in Fig. 2 to eliminate the ISI?
- iii) If both of these two problems are solved, how does the receiver recover the input signal x[n] from the received $\hat{x}[n]$?

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equivalent to the fractionally spaced equalizer studi example, [36]–[39], where the receiver needs to sat signal N times faster than the baud sampling. When L the precoding concept has appeared in [39] by Tsatsat Giannakis, where the precoder $G_l(z) = c_l, l = 0, 1, \cdots$ for N constants c_l was used. As we can see, the c_l K = 1 is a very special case in our precoding scheme moreover, our new precoding scheme in Fig. 2 provide potential precoders $G_l(z), l = 0, 1, \cdots, N - 1$ rathe only constants c_l , which allows one to search the optim with respect to an individual channel.

When $K \ge N$ and there are N interference channels of a single channel H(z) in Fig. 2, a detailed analys given by Nguyen [31]. When K > N, as mention [31], PR is impossible, but partial alias cancellation filte were proposed in [31]. The applications discussed in [in wide-band radio communications, where only part signal frequencies is of interest to the user. In this we are interested in applications in the ISI channels w systems in Fig. 2 and, therefore, the case of K < Nalso implies that unlike the existing precoding technique new precoding expands the transmission bandwidth, w what we lose for the new precoding method, and fortuwe will show that the bandwidth expansion can be as as possible in theory.

An intuitive way to reduce the ISI generated from a l H(z) is to smoothly interpolate x[n] with a large number of interpolations between samples of x[n]the interpolated one has the lowpass property. Howev drawbacks about this approach may occur. One is usually requires a large amount of increasing of da (number of interpolations between samples). The c that a good frequency band structure for a nonlowpas as bandpass, filter H(z) is required for PR. In this we want to solve the above three problems systematic Given two integers 0 < K < N, we present a ne and sufficient condition (see Theorem 1) on an FII H(z) such that there exists an FIR nonmaximally dec multirate filterbank with N channels and decimation b that x[n] can be recovered from $\hat{x}[n]$ in Fig. 2 with synthesis bank. The condition we found is basically very In fact, it can be proved that for any given FIR filte not identically 0, there always exists an FIR nonmax decimated multirate filterbank in Fig. 2 for recovering

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