Adaptive Modulation for SFBC-OFDM Systems with Zero-Forcing Equalization

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ABSTRACT
The Space frequency block coded OFDM gives better performance in highly frequency selective and time varying channel than STBC OFDM. Hence this system can be mostly useful in vehicular communication where user is highly mobile. In this paper, An Adaptive modulation scheme for SFBC-OFDM system is proposed to give the reliable connectivity as well as efficient data rate to highly mobile user. Here, the bit rate is adapted according to the SNR value estimated from channel. For variation in bit rate the M-QAM modulations scheme is selected by comparing estimated SNR value and threshold value for switching. Also to reduce the effect channel which distorted the transmitted signal the Zero forcing (ZF) equalization is used. Finally performance of proposed system is analyzed with BER VS SNR plot.

Keywords
Orthogonal Frequency Division Multiplexing (OFDM), Space Frequency Block coding (SFBC), Space time block coding (STBC)

1. INTRODUCTION
Feature mobile communication must satisfies the growing demands of High data rate, reliable communication, low power consumption as well as spectral efficiency. In case of vehicular mobile communication, where the high mobility scenario occurs the wireless connection is not trusted or it fails to give expected data rates. In case of high mobility scenario, the received signal quality degrades to more extent due to the inter carrier interference (ICI) and inter symbol interference (ISI) problems. It happens because of time variation in channel due to mobile speed characterized by Doppler spread. Also antenna selection is the bigger problem in high speed vehicular communication. As the channel is frequency selective or fast fading the diversity is the best solution to the fast fading channels. Higher the diversity gain, lower the effect of ISI and ICI on the transmitted signal. Aloumoti STBC give better diversity gains[1]. Hence they play important role in today’s wireless communication system along with OFDM. In STBC OFDM the replica of the signal is transmitted on the same subcarrier of two consecutive OFDM symbols. Hence, STBC gives better performance where the channel remains constant across two OFDM symbols duration. Hence STBC OFDM fails to give reliable connection. In this case SFBC OFDM gives better performance as in SFBC the replica of signal is transmitted on neighboring subcarrier of same OFDM symbol. It is assumed that channel remains constant across two consecutive subcarriers [3]. Along with the diversity there is need of good signal detector at the receiver. For receive diversity simple maximum-likelihood (SML) detector. However, this detector is optimum in quasi static channel only. Recently, Vielmonet al. recommended three novel detectors for time varying channel to combat the effect of rapid channel variation. These detectors are zero forcing (ZF), joint maximum-likelihood (JML) and decision-feed back (DF). Out of this the DF is used in this paper because of its low complexity [4].

A major drawback of the wireless vehicular communication system is deep fading due to high mobility of device. In SFBC OFDM the fixed modulation scheme is used which fails to mitigate the deep fading problem. These problems can be minimized by adapting the parameters like modulation scheme, power, and ranges of system parameters based on channel condition. In 1968 Hayes proposed power adaptive modulation scheme [5]. Adaptive scheme based on mode switching level is proposed by Hanzo et al [6-9] shows that it provides better spectral efficiency and BER performance than fixed modulation. For multcarrier system the adaptive modulation is performed by Klaet known as AOFDM [10]. Advantages of diversity technique along with AOFDM are investigated in [11]. Hence, in this paper an adaptive modulation technique is proposed for SFBC OFDM system. In which the modulation schemes assigned to SFBC OFDM system are changed according to channel conditions. Channel condition is measured in terms of received SNR estimated from channel noise and AWGN noise. Also ZF detection technique is applied at receiver to retrieve the original signal. Finally the performance of the proposed adaptive modulation system is analyzed in terms of SNR VS BER plots. Also, performance is given for different detection schemes as mentioned above.

So this paper is arranged in following sections section 2 gives the Overview of SFBC OFDM SYSTEM with simple SFBC encoding-decoding. In Section 3 review of signal equalization technique at receiver. Section 4 describes proposed adaptive modulation technique for SFBC-OFDM. In Section 5 the performance of SFBC OFDM system is analyzed with and without adaptive modulation Section 6 concludes this paper.

2. SFBC-OFDM SYSTEM MODEL
In Space time block coding, the symbols of an orthogonal design are transmitting on same subcarrier of subsequent OFDM symbol due to which performance of system degrades in fast channel variation in time. Hence, space frequency block coding is preferred to avoid problem of fast channel variation in time. In SFBC the symbols of
orthogonal design transmitted on neighboring Subcarrier of same OFDM symbol. This is will illustrate in Figure1.

\[
\begin{bmatrix}
X_1 \\
X_2
\end{bmatrix} \rightarrow \begin{bmatrix}
X_1 \\
-X_2^* \\
X_1^*
\end{bmatrix} \rightarrow \text{space 1 subcarrier}
\]

Fig.1: Alamouti SFBC[2]

The example in Fig 2, show the basic idea behind the implementation of SFBC-OFDM system.

Fig.2: SFBC in OFDM With two Transmitters [2]

To examine SFBC-OFDM in more depth let us consider that \(x_i(n)\) is \(n^2\) signal transmitted from ith antenna then \(x_i(n)\) is given as:

\[
x_i(n) = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} x_i(k) W^{-nk}_N
\]

\[
n = 0,1...N-1
\]

Where, \(W^{-nk}_N\) is twiddle factor, \(x_i(n)\) is original QPSK signal to be transmitted Equation (1) can be written as

\[
x_i(n) = \frac{1}{\sqrt{N}} \sum_{m=0}^{N-1} [x_i(2m) + W^{-n}x_i(2m+1)]W_{nm}\]

\[
= \frac{1}{\sqrt{2}}[x^n(n) + W^{-n}x^o(n)](2)
\]

Where, \(x^n(n)\) points IFFT of even samples of \(x_i(n)\) and \(x^o(n)\) corresponds to odd samples of \(x_i(n)\).

Similarly for second antenna

\[
x_2(n) = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} x_2(k) W^{-nk}_N
\]

\[
x_2(n) = \frac{1}{\sqrt{N}} \sum_{m=0}^{N-1} [x_2(2m) + W^{-n}x_2(2m+1)]W_{nm}\]

Now applying Alamouti scheme to adjacent carriers we have

\[
X_1(n) = [X_1(0), X_1(1), X_1(2) ... X_1(N-1)]]
\]

\[
X_2(n) = [-X_1(1), X_1(0), -X_1(3) ... X_1(N-1)]]
\]

\[
x_2(n) = \frac{1}{\sqrt{N}} \sum_{m=0}^{N-1} [-X_1(2m+1) + W^{-n}X_1(2m)]W_{nm}\]

\[
= \frac{1}{\sqrt{2}}[x^e(n) + W^{-n}x^o(n)]
\]

CP is appended at the head of each symbol to eliminate ISI making all channel matrices circulant \((H_i)\). Received symbol after removal of CP is given by

\[
y = \frac{1}{\sqrt{2}}[H_1x_1 + H_2x_2 + n]
\]

Similarly, if two transmitter and receivers are used then eq(3) becomes

\[
\begin{bmatrix}
y_1 \\
y_2
\end{bmatrix} = \frac{1}{\sqrt{2}}\begin{bmatrix}
h_{11} & -h_{12} \\
h_{21} & h_{22}
\end{bmatrix}\begin{bmatrix}
x_1 \\
x_2
\end{bmatrix} + \begin{bmatrix}
{n_1} \\
n_2
\end{bmatrix}
\]

\[
(4)
\]

In equation (2) \(h_{ij}, i,j \in \{1,2\}\) denotes the complex channel response at OFDM subcarrier i from transmit antenna j,x1 and \(n_2\)are independent identically distributed (i.i.d) zero-mean complex Gaussian Circularly symmetric with variance \(\sigma^2\) that represents the additive noise term. Each subcarrier assumed to experiences the frequency flat Rayleigh fading.

Now estimated channel response can be modeled as

\[
\hat{H} = H + \begin{bmatrix}
e_{11} \\
e_{22}
\end{bmatrix}
\]

\[
(5)
\]

Where, E is random matrix where elements \(e_{ij}\) are i.i.d zero-mean complex Gaussian Circularly symmetric with variance \(\sigma^2\)

At, receiver side the linear decoding of SFBC signal is done as [12]

\[
z = \frac{1}{\sqrt{2}}H^H y
\]

\[
= \frac{1}{\sqrt{2}}(H^HE^H)\left(\frac{1}{\sqrt{2}}Hx + n\right)
\]

\[
= \frac{1}{2}\begin{bmatrix}
h_{11}^2 + |h_{22}|^2 & h_{12}^2 - h_{11}^2 & h_{11}^2 + h_{12}^2 \end{bmatrix}_x
\]

\[
+ \frac{1}{2}E^HHx + \frac{1}{\sqrt{2}}(H^HE^H)n
\]

\[
(5a)
\]

\[
= \frac{1}{2}\begin{bmatrix}
h_{11}^2 + |h_{22}|^2 & x_1 \\
|h_{22}|^2 & h_{11}^2 + h_{12}^2\end{bmatrix}
\]

\[
+ \frac{1}{2}E^HHx + \frac{1}{\sqrt{2}}(H^HE^H)n
\]

\[
(5b)
\]

\[
(5c)
\]

Where \(A^H\) is hermitian transpose of A. The term in (5a) is desired component, (5b) leads to the self interference terms and (5c) denotes the additive noise terms.

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3. EQUALIZATION TECHNIQUE

In MIMO the ISI occurs due to the multipath propagation. This distorted the transmitted signal which in turn causes bit error rate at receiver. Hence, equalizer is used to minimize the error between actual output and desired output by continuous updating its filter coefficient. Equalization can be used in both frequency and time domain. In this paper Zero forcing equalization method is implemented in frequency domain.
3.1 Zero forcing equalization

In Zero-forcing equalization the coefficient are chosen to force the samples of the combined channel and equalizer impulse response to zero. [13] The combined impulse response of the channel is given by

\[ H_{\text{ch}}(f)H_{\text{eq}}(f) = 1 \]  

(6)

Where, \( H_{\text{ch}}(f) \) is folded frequency response of channel and \( H_{\text{eq}}(f) \) is frequency response of equalizer. Zero forcing equalizer can be realized by multiplying equation (5b) by vector \( 1/H(K) \) where \( H(K) \) is normalized MIMO-channel vector which can be formed as:

\[ H = h_{12}^*h_{22} - h_{11}h_{22} \]  

(7)

In this case the equalization filter compensates for the channel induced ICI as well as ISI brought about by the transmitter and receiver filters. Zero forcing equalizer designed above does not eliminate all ISI because the filter is of finite length.

4. PROPOSED SCHEME ADAPTIVE MODULATION FOR SFBC-OFDM

The allocation of bits to the subcarriers SFBC-OFDM system can be uniform or non uniform. In frequency selective fading channel some subcarriers undergo deep fade while some undergo less fading. The occurrence of bit errors is due to the deep fading of subchannels. To minimize the occurrence of bit errors at deeply faded subcarriers, one can change the bits allocated to those subcarriers i.e allocate less bits than if uniform. This change can be done by changing the modulation scheme used for transmission [14]. In this system 8QAM, 16QAM, 32QAM and 64QAM modulation schemes are used. The change in modulation scheme depends on the estimated value of SNR from channel at receiver and threshold value for switching. The SNR estimation at receiver is given as

\[ \gamma_0 = \frac{P_t}{N_0/2} \]  

(9)

Where, \( \sigma_n^2 = N_0 \) under the assumption of circular symmetric complex Gaussian random variables and \( N_0/2 \) is the power spectral density of real/imaginary component of noise. Fig 3 show the block schematic for proposed SFBC-OFDM system with adaptive modulation.

One can also detect the SNR value by computing the power of wanted signal divided by the power of noise. But from this SNR before noise reduction is computed. To get the SNR after noise reduction, Residual noise power needs to be computed and it is found as follows.

SNR before noise reduction = \[ \sum \frac{\text{power of wanted signal}}{\text{power of noise}} \]  

Residual noise = Actual signal – noise reduced signal

SNR after noise reduction = \[ \sum \frac{\text{power of wanted signal}}{\text{power of residual noise}} \]

(10)

(11)

The estimated value of SNR in (8) or (11) at receiver is used for selection of modulation scheme at transmitter. This estimated SNR at receiver is feedback to transmitter. At transmitter the estimated SNR and thresholds for selection are compared and based on this comparison the modulation scheme is selected. Table 1 shows the SNR value range and respective thresholds for modulation scheme selection.

The capacity of the channel is estimated using the below equation

\[ C = \log_2(1 + \text{SNR}) \]  

(12)

Where B is the bandwidth of transmitted signal and SNR is the estimated SNR of received signal [16].

<table>
<thead>
<tr>
<th>MODULATION</th>
<th>THRESHOLD</th>
</tr>
</thead>
<tbody>
<tr>
<td>4QAM</td>
<td>0 ≤ SNR &lt; 6</td>
</tr>
<tr>
<td>8QAM</td>
<td>6 ≤ SNR &lt; 10.5</td>
</tr>
<tr>
<td>16QAM</td>
<td>10.5 ≤ SNR &lt; 13.5</td>
</tr>
<tr>
<td>32QAM</td>
<td>13.5 ≤ SNR &lt; 16.6</td>
</tr>
<tr>
<td>64QAM</td>
<td>16.6 ≤ SNR</td>
</tr>
</tbody>
</table>

5. SIMULATION AND DISCUSSION

At transmitter initially input signal is 32QAM modulated and passed through SFBC-OFDM system. Then output of SFBC-OFDM system transmitted on frequency-selective Rayleigh fading channel. At the receiver, the strength of received signal is calculated and channel estimation is done. Zero forcing Channel equalization method is used to remove the effect of ISI. The received SNR is calculated using the signal strength and channel estimation. The estimate selects the appropriate constellation size of M-QAM signaling that suits the channel condition. The data rate is adapted by the transmitter by changing modulation level based on the received SNR. The flow chart is for the simulation is shown in fig (5).
5.1 SIMULATION PARAMETERS
Table (2) gives the simulation parameters which are taken from slandered wireless data. The simulation is done as per the flow chart shown in fig (3). At first the bit stream for transmission is generated. Then this stream is modulated with M-QAM modulation scheme and symbols are formed. These symbols are encoded using SFBC encoder. Then OFDM modulation is performed on each SFBC encoded stream. Finally generated OFDM symbols are transmitted via frequency selective Rayleigh fading channel. At receiver the SFBC encoded signal is detected using the Zero forcing detector. Then it is demodulated and decoded to get original signal back. BER and SNR are calculated.

Table 2: Simulation parameters and Constraints

<table>
<thead>
<tr>
<th>Parameters</th>
<th>Type</th>
</tr>
</thead>
<tbody>
<tr>
<td>Modulation</td>
<td>M-QAM (M=4,8,16,32,64)</td>
</tr>
<tr>
<td>MIMO antenna configuration</td>
<td>2x2</td>
</tr>
<tr>
<td>Signal to noise ration</td>
<td>0:5:40 dB</td>
</tr>
<tr>
<td>Average transmit power</td>
<td>1db</td>
</tr>
<tr>
<td>Fading</td>
<td>Frequency selective Rayleigh fading</td>
</tr>
<tr>
<td>Carrier frequency</td>
<td>900 MHz</td>
</tr>
<tr>
<td>Bandwidth of signal(Bs)</td>
<td>200MHz</td>
</tr>
<tr>
<td>Symbol period</td>
<td>1/Bs</td>
</tr>
<tr>
<td>NO of paths</td>
<td>12</td>
</tr>
</tbody>
</table>

5.2 RESULTS AND DISCUSSION
In analysis SFBC-OFDM system with adaptive modulation. The results are derived for Channel capacity vs SNR and BER vs SNR. Fig 4 gives the capacity of channels for 8 QAM, 16 QAM, 32 QAM and 64 QAM modulations under frequency selective Rayleigh fading channel. Fig 6 gives the capacity of channel for the adaptive modulation technique.
6. CONCLUSION

In this paper the capacity vs SNR and BER vs SNR performance of proposed adaptive modulation scheme for SFBC-OFDM scheme is analyzed. For lower SNR values the capacity of channel is very low. Hence lower order modulation scheme like 8 QAM gives less BER for Low SNR values. Hence it helps in maintaining the connectivity in very bad channel conditions. But when channel conditions are good then received SNR values are high. At this time the capacity of channel is also high. Here 8QAM again gives very low BER but data rate is not achieved as channel allows more data rate. So for achieving data rate in good channel conditions the higher modulation order are used.

Adaptive modulation gives the average performance which helps in achieving data rate as well as reliable connectivity.

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7. REFERENCES


