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- (71) Applicant (for all designated States except US): TEXAS INSTRUMENTS INCORPORATED [US/US]; P.O. Box 655474, Mail Station 3999, Dallas, TX 75265-5474 (US).
- (72) Inventors; and
- (75) Inventors/Applicants (for US only): DABAK, Anand, G. [US/US]; 7700 Cherry Creek Drive, Plano, TX 75025 (US). ONGGOSANUSI, Eko, N. [ID/US]; 1430 McClure Drive, Allen, TX 75013 (US).
- (74) Agents: FRANZ, Warren, L. et al.; Texas Instruments Incorporated, Deputy General Patent Counsel, P.O. Box 655474, Mail Station 3999, Dallas, TX 75265-5474 (US).

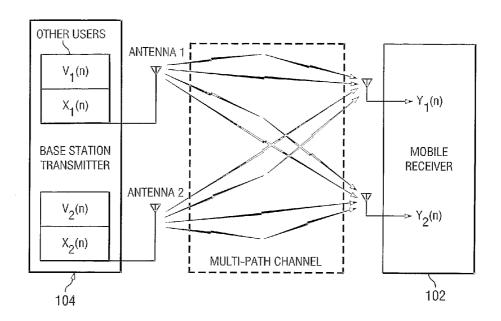
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[Continued on next page]

(54) Title: FREQUENCY DOMAIN EQUALIZATION



(57) Abstract: A radio receiver 102 is provided. The radio receiver 102 comprises one or more data Fast Fourier Transformers, each data Fast Fourier Transformer operable to perform a Fast Fourier Transformers, each impulse response Fast Fourier Transformer operable to perform a Fast Fourier Transform on a channel impulse response, one or more multiplier components operable to multiply a term of the output of one of the data Fast Fourier Transformers by a term of the output of one of the impulse response Fast Fourier Transformers, and one or more Inverse Fast Fourier Transformers, each Inverse Fast Fourier Transformer operable to perform an Inverse Fast Fourier Transform based on an output of one or more of the multipliers.



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FREQUENCY DOMAIN EQUALIZATION

[0001] This invention relates in general to wireless communications; and, more specifically, but not by way of limitation, to frequency domain equalization.

BACKGROUND OF THE INVENTION

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[0002] The downlink of high speed downlink packet access (HSDPA) for third generation partnership project (3GPP) compliant systems tends to suffer from multi-path, and requires the use of a chip level equalizer instead of a conventional rake receiver architecture used in communication systems, such as code division multiple access (CDMA) communication systems. Chip level equalization for HSDPA is typically performed using a time domain filter computation based upon various techniques such as normalized least mean square (NLMS), two-stage NLMS, Griffith's, pre-filter rake, channel estimation based LMS (CHEST), Fast Fourier Transform (FFT) based minimum mean squared error (MMSE) filter calculation or channel response based minimum output energy (CR-MOE).

[0003] HSDPA delivers superior speed, capacity and efficiency improvements. Some advantages to HSDPA include improved spectrum efficiency, improved network capacity, higher peak data rates, and improved allocation of resources.

SUMMARY OF THE INVENTION

[0004] A radio receiver is provided. A described radio receiver comprises: one or more data Fast Fourier Transformers, each data Fast Fourier Transformer operable to perform a Fast Fourier Transform on an input data block; one or more impulse response Fast Fourier Transformers, each impulse response Fast Fourier Transformer operable to perform a Fast Fourier Transform on a channel impulse response; one or more multiplier components operable to multiply a term of the output of one of the data Fast Fourier Transformers by a term of the output of one of the impulse response Fast Fourier Transformers; and one or more Inverse Fast Fourier Transformers, each Inverse Fast Fourier Transformer operable to perform an Inverse Fast Fourier Transform based on an output of one or more of the multipliers.

[0005] A method for performing frequency domain equalization on data is also provided. A described method comprises: (a) receiving data that needs to be demodulated, the data including a first vector Y_1 and a first channel impulse response, H_1 ; (b) performing Fast Fourier Transforms (FFT) on the first vector Y_1 and on the first channel impulse response H_1 ; (c) multiplying the data

by a term of the FFTs of the first vector Y_1 and the first channel impulse response H_1 ; and (d) performing an Inverse FFT (IFFT) on the result of (c).

A method for performing frequency domain equalization on data is also provided. A [0006] described method comprises: (a) receiving a first vector Y₁, a first channel impulse response H₁₁, and a second channel impulse response H₂₁; (b) receiving a second vector Y₂, a third channel impulse response H₁₂, and a fourth channel impulse response H₂₂; (c) performing Fast Fourier Transforms (FFT) on the first vector Y1, the second vector Y2, the first channel impulse response H₁₁, the second channel impulse response H₂₁, the third channel impulse response H₁₂, and the fourth impulse response H₂₂; (d) determining a first product by multiplying the FFT of the first vector Y_1 by the FFT of the first channel impulse response H_{11} , a second product by multiplying the FFT of the first vector Y_1 by the FFT of the second channel impulse response H_{21} , a third product by multiplying the FFT of the second vector Y₂ by the FFT of the third channel impulse response H₁₂, and a fourth product by multiplying the FFT of the second vector Y2 by the FFT of the fourth impulse response H₂₂; (e) determining a first sum based on summing the first, second, third, and fourth products; (f) determining a first output based on performing an Inverse FFT (IFFT) of the first sum; (g) determining a second sum based on summing the first, second, third, and fourth products; and (h) determining a second output based on performing an IFFT of the second sum.

BRIEF DESCRIPTION OF THE DRAWINGS

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[0007] FIG. 1 is a diagram of a multiple transmit and receive antenna system for HSDPA in accordance with an embodiment of the invention.

[0008] FIG. 2 shows a diagram highlighting the relationship between the block of interest and a block used in FDE in accordance with an embodiment of the invention.

[0009] FIG. 3 shows the new blocks used in demodulation by overlapping with the previous blocks shown in FIG. 2 in accordance with an embodiment of the invention.

25 **[0010]** FIG. 4 shows a flowchart for performing the frequency domain equalization in accordance with an embodiment of the invention.

[0011] FIG. 5 shows a diagram of a radio in accordance with an embodiment of the invention.

[0012] FIG. 6 shows a block diagram for a one transmit and one receive antenna system in accordance with an embodiment of the invention.

30 [0013] FIG. 7 shows a diagram of the data used at the despreader input in accordance with an embodiment of the invention.

[0014] FIG. 8 shows a diagram of a frequency domain equalizer for a one transmit/two receive antenna system in accordance with an embodiment of the invention.

[0015] FIG. 9 shows a diagram for a frequency domain equalizer for a two transmit/one receive antenna system in accordance with an embodiment of the invention.

5 [0016] FIG. 10 shows a diagram for a frequency domain equalizer for a two transmit/two receive antenna system in accordance with an embodiment of the invention.

[0017] FIG. 11 shows a block diagram of a frequency domain equalizer architecture in accordance with an embodiment of the invention.

DETAILED DESCRIPTION OF EMBODIMENTS

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[0018] In one embodiment of the invention, a Fast Fourier Transform (FFT) based chip level equalization is described. The equalization does not require time domain filter computation of the chip level equalizer (CLE) as prior approaches. The equalization is performed in the frequency domain using in one illustrative example, a well-known digital signal processing (DSP) technique such as the windowing, overlap and save technique. The estimated complexity for the resulting chip level equalizer is expected to be about the same as previous FFT based chip level equalization computing time domain filters.

[0019] FIG. 1 shows a multiple transmit and receive antenna system for HSDPA in accordance with an embodiment of the invention. A base station 104 having two antennas is shown communicating with a mobile receiver 102 having two antennas via a multi-path channel. An FFT-based frequency domain equalization (FDE) for various antenna-setups, such as 1x1, 1x2, 2x1, 2x2 are also described in order to better understand the invention. Extension to systems with more antennas at the transmitter and/or receiver than those described is also supported by the present invention by making simple adjustments.

One Transmit And One Receive Antenna System Case:

[0020] If there is only a single transmit and a single receive antenna in the HSDPA system, the channel impulse response is assumed to be given by $h^T = \{h_1, h_2,, h_N\}$. In the case of pedestrian A, pedestrian B, vehicular A channels found in reality, the various intermediate taps may be zero, but the receiver is assumed to estimate the channel for the full span for the channel response. The span of the channel response may be indicated by the 1999 release of the HSDPA modem standard and this is a long term non-real time phenomenon. The received data 202 and a block of interest for FDE 204 are shown in FIG.2.

[0021] As shown in FIG. 2, the block to be used in FDE is given by $Y^T = \{y_1, y_2, y_3, ..., y_M\}$ and the block of interest for actual demodulation 204 is a subset of the above block Y and is given by $\widetilde{Y}^T = \{y_k, y_{k+1}, ..., y_{M-k+1}\}$ where k > 0. The received data vector Y can now be written as:

$$Y = \begin{bmatrix} h_0 & & & & & \\ h_1 & h_0 & & & & \\ & h_1 & h_0 & & & \\ & & h_1 & & & \\ & & & h_1 & & \\ & & & h_N & & & h_0 & \\ & & & h_N & & & h_1 & h_0 \\ & & & & h_2 & h_1 & h_0 \end{bmatrix} \begin{bmatrix} x_0 \\ x_1 \\ x_2 \\ & & \\ & & \\ & & \\ &$$

[0013] The channel matrix H is a Toeplitz matrix of dimension $M \times M$. We now propose to make the matrix H circulant by copying certain terms of the channel response in the North East (top right) corner of the matrix H as given in the equation (2) below and call the resulting receive vector as Y_{circ} as given in equation (3) below:

$$H_{circ} = \begin{bmatrix} h_0 & & & & h_2 & h_1 \\ h_1 & h_0 & & & h_N & & h_2 \\ & h_1 & h_0 & & & h_N & \\ & & h_1 & & & & h_N \\ & & & h_1 & & & & h_N \\ & & & h_N & & & h_0 & \\ & & & h_N & & & h_0 & \\ & & & & h_N & & & h_0 & \\ & & & & h_N & & & h_1 & h_0 \\ & & & & & h_2 & h_1 & h_0 \end{bmatrix}$$
 Equation (2), and

 $Y_{circ} = H_{circ}X + N$. Equation (3)

[0014] Since H_{circ} is a circulant matrix, it can be represented as $H_{circ} = F^H \Lambda F$ where Λ is diagonal matrix and F is an FFT matrix. Using this in equation (3) the result is;

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$$Y_{circ} = F^H \Lambda F X + N \Longrightarrow F Y_{circ} = \Lambda F X + F N \; .$$

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Since Λ is a diagonal matrix, the MMSE solution for X can now be seen to be

$$\hat{X}_{MMSE} = F^{H} \left(\left[\Lambda^{H} \Lambda + \sigma^{2} I \right]^{-1} \Lambda^{H} F Y_{circ} \right).$$
 Equation (4)

[0015] Since A is a diagonal matrix, the matrix inverse in the above equation is a scalar inversion of the individual elements along the diagonal. The \hat{X}_{MMSE} is the estimate of the transmitted signal over the whole block (A) shown in FIG. 2. Because of the approximation in equation (2) of making the matrix H being circulant, the estimates of the transmitted signal towards the edge of the block A will not be reliable. Hence, only the estimates of X for a block of data (B) shown in FIG. 2 are considered and the remaining data from the block (A) is discarded. Next, in order to receive the remaining data in block (A) 202 the procedure is repeated where in another overlapping block A' 304 is formed as shown in FIG. 3. Block 302 is the new block used in the frequency domain equalization, with 304 (B') is the new block of interest for demodulation.

10 [0016] Referring now to FIG. 4, there is shown a flowchart highlighting the frequency domain equalization for HDSPA in accordance with one embodiment of the invention. In 402, block for demodulation (B) 204 from the received data is selected. A larger block (A) 202 is picked that overlaps (B) 204, received data Y. In 406, the FFT of vector Y and channel impulse response H are taken. In 408, the term is multiplied by the term of the two FFT's and normalized using equation (4). In 410, an inverse fast fourier transform (IFFT) of the resultant signal is computed. The edges of the resultant data are truncated in 412 in order to get the chip level estimates for the transmitted data for the block (B) 204. Finally, in 414, the data is despread in order to produce symbol level estimates. With judicious choice of equalizer parameters (e.g., the amount of overlapping, block size), it is expected that the FFT-based FDE performs at least as well as the time-domain counterparts (e.g. NLMS, Griffith, FFT-based TD-MMSE).

[0017] In FIG. 5, there is shown a diagram of a radio 502 that includes a receiver that implements the FDE of the present invention. In one illustrative example, radio 502 supports HSDPA.

Complexity estimates:

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25 [0018] Complexity estimates for the FDE for HSDPA for typical values of the lengths for blocks A, B are now described. For a length, L, the total number of complex multiplies are FFT'd if M/2log2(M) and M/2log2(M) are complex additions. The total of operations per chip is given in table 1 below, for the FDE for different chosen lengths for blocks A 202, B 204 from FIG. 2. The length of block A 202 is denoted as M, and that for block B is denoted as, P.

30 **[0019]** For comparison purposes to the present invention, consider a basic NLMS time domain equalizer with length 16 taps filter. Using the NLMS equalizer, the number of operations needed per

chip are; 16 complex multiplies per chip for the filtering, plus 16 complex multiplies per chip, for the NLMS update, implying a total of at least 32 complex multiplies per chip. Similarly, the number of complex additions per chip are 15 complex adds for filtering, plus 15 complex adds for the LMS update. This implies a total of 30 complex additions per chip.

[0020] Thus, the complexity of the frequency domain based equalizer of the present invention is less than a single stage of the NLMS time domain equalizer, while achieving practically the performance of an MMSE equalizer. It should be noted that the complexity estimate for the FDE does not take the channel estimation complexity into account.

Table 1

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	M = 256, P = 128	M = 128, P = 64	M = 64, P = 16	M = 32, P = 16
FFT(Y)	1024 X and +	448 X and +	192 X and +	80 X and +
FFT(h) with zero	1024 X and +	448 X and +	192 X and +	80 X and +
padding				
Z = FFT(Y)*FFT(h)	256 X	128 X	64 X	32 X
$z1 = z / \left(\left[\Lambda^H \Lambda + \sigma^2 I \right] \right)^{-1}$	512 X, 256 +	256 X, 128 +	128 X, 64 +	64 X, 32 +
IFFT(Z)	1024 X and +	448 X and +	192 X and +	80 X and +
Total	3840 complex X,	1728 complex X,	768 complex X,	304 complex X,
	3584 complex +	1600 complex +	704 complex +	272 complex +
Operations/chip = Total/P	30 complex X,	27 complex X,	48 complex X,	19 complex X,
	28 complex +	25 complex +	44 complex +	17 complex +

FDE as a rake receiver:

[0021] It is very important that the equalizer be able to do simple matched filtering operations and behave like a rake receiver so that the performance of the equalizer may not degrade compared to a rake receiver. In equation (4) this can be done simply by not taking the term $\Lambda^H \Lambda$ into account during the equalization phase. Thus, whenever the receiver is to be forced to behave as a rake receiver the equation becomes:

$$\hat{X}_{rake} = F^H \Lambda^H F Y_{circ} .$$
 Equation (5)

One transmit, two receive antenna:

[0022] Similar to FIG. 2, we again consider a block of interest for demodulation (B) from the two receive antennas and a larger sized block (A) for the block used in FDE. Skipping some of the obvious details, let us start by writing the equivalent equation (3) for the 1X2 antenna case;

$$\begin{bmatrix} Y_{circ}^{1} \\ Y_{circ}^{2} \end{bmatrix}_{2MYM} = \begin{bmatrix} H_{circ}^{1} \\ H_{circ}^{2} \end{bmatrix}_{2MYM} X_{MXM} + \begin{bmatrix} N^{1} \\ N^{2} \end{bmatrix}_{2MYM}$$
Equation (6)

Where the superscripts indicate the receive antenna. In this case, once exploiting the property of circulant matrices, we get;

$$\begin{bmatrix} Y_{circ}^{1} \\ Y_{circ}^{2} \end{bmatrix}_{2MXM} = \begin{bmatrix} F^{H} & 0 \\ 0 & F^{H} \end{bmatrix}_{2MX2M} \begin{bmatrix} \Lambda^{1} \\ \Lambda^{2} \end{bmatrix}_{2MXM} F_{MXM} X_{MXM} + \begin{bmatrix} N^{1} \\ N^{2} \end{bmatrix}_{2MXM} = >$$

$$10 \quad \begin{bmatrix} F & 0 \\ 0 & F \end{bmatrix} \begin{bmatrix} Y_{circ}^{1} \\ Y_{circ}^{2} \end{bmatrix}_{2MXM} = \begin{bmatrix} \Lambda^{1} \\ \Lambda^{2} \end{bmatrix}_{2MXM} F_{MXM} X_{MXM} + \begin{bmatrix} N^{1} \\ N^{2} \end{bmatrix}_{2MXM} = >$$

$$\hat{X}_{MMSE} = F_{MXM} \begin{bmatrix} \Lambda^{1H} \Lambda^{1} + \Lambda^{2H} \Lambda^{2} + \sigma^{2}I \end{bmatrix}^{-1} MXM \begin{pmatrix} \Lambda^{1H} FY_{circ}^{1} + \Lambda^{2H} FY_{circ}^{2} \end{pmatrix}$$
Equation (7)

[0023] The receiver performs a similar process to that shown in FIG. 4. In terms of complexity, looking at equation (7) one can note that the complexity for the 1X2 FDE case is less than twice the complexity of the single FDE since there is only one final IFFT. Hence, the expected complexity of the 1X2 FDE case is less than or comparable to a one stage NLMS, not including the channel estimation complexity for the FDE.

Two transmit, one receive antenna case:

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[0024] Similar to FIG. 2, a block of interest for demodulation (B) from the two receive antennas and a larger sized block (A) for the block used in FDE is considered. Starting by writing the equivalent equation (3) for the 2X1 antenna case:

$$Y_{circ} = \begin{bmatrix} H_{circ}^{1} & H_{circ}^{2} \end{bmatrix}_{MX2M} \begin{bmatrix} X^{1} \\ X^{2} \end{bmatrix}_{2MYM} + N_{MXM}$$
 Equation (8)

Where the superscripts indicate the transmit antenna. In this case, exploiting the property of circulant matrices we get:

$$Y_{circ} = F^{H} \begin{bmatrix} \Lambda_{circ}^{1} & \Lambda_{circ}^{2} \begin{bmatrix} F & 0 \\ 0 & F \end{bmatrix} \begin{bmatrix} X^{1} \\ X^{2} \end{bmatrix} + N = >$$

$$FY_{circ} = \begin{bmatrix} \Lambda_{circ}^{1} & \Lambda_{circ}^{2} \begin{bmatrix} F & 0 \\ 0 & F \end{bmatrix} \begin{bmatrix} X^{1} \\ X^{2} \end{bmatrix} + FN = >$$
Equation (9)

For a given Y = HX + N, the MMSE equalizer is given two equivalent formulations namely; $(H^H H + \sigma^2 I)^{-1} H^H = H^H (HH^H + \sigma^2 I)^{-1}$. Now using the second formulation in equation (9) we now get:

$$\hat{X}_{MMSE} = \begin{bmatrix} F^{H} & 0 \\ 0 & F^{H} \end{bmatrix}_{2MY2M} \begin{bmatrix} \Lambda^{1H} \\ \Lambda^{2H} \end{bmatrix}_{2MYM} \left(\Lambda^{1}_{circ} \Lambda^{1H}_{circ} + \Lambda^{2}_{circ} \Lambda^{2H}_{circ} + \sigma^{2} I \right)^{-1} FY_{circ}$$
 Equation (10)

[0025] Since the matrix inverse in the above equation is indeed a scalar inverse for each of the frequency terms, it can be interchanged with the rest of the operations. Comparing equation (10) to equation (7), it can be seen that structurally the two equations are similar, hence similar hardware architecture implementation for 1X2 and 2X1 can be used in practice. The receiver performs a similar process as shown in FIG. 4. In terms of complexity, looking at equation (10) it can be seen that the complexity for the 2X1 FDE case is less than twice the complexity of a single FDE since there is only 1 FFT. Hence, the expected complexity of the 2X1 FDE case is less than or comparable to the 1 stage NLMS case, not including the channel estimation complexity for the FDE.

Two transmit, two receive antenna case:

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[0026] Similar to FIG. 2, we consider a block of interest for demodulation (B) from the two receive antennas and a larger sized block (A) for the block used in FDE. Skipping some of the basic steps, we start by writing the equivalent equation (3) for the 2X2 case:

$$\begin{bmatrix} Y_{circ}^1 \\ Y_{circ}^2 \end{bmatrix} = \begin{bmatrix} H_{circ}^{11} & H_{circ}^{21} \\ H_{circ}^{12} & H_{circ}^{22} \end{bmatrix}_{2MX2M} \begin{bmatrix} X^1 \\ X^2 \end{bmatrix}_{2MXM} + \begin{bmatrix} N^1 \\ N^2 \end{bmatrix}_{2MXM}$$
 Equation (11)

where superscript ij for the channel matrix implies the channel from transmit antenna i to receive antenna j. Similarly, superscripts on Y indicate the receive antenna and the superscript on X indicates the data on the transmit antenna. Exploiting the circulant property the equation becomes:

$$\begin{bmatrix} Y_{circ}^{1} \\ Y_{circ}^{2} \end{bmatrix} = \begin{bmatrix} F^{H} & 0 \\ 0 & F^{H} \end{bmatrix} \begin{bmatrix} \Lambda^{11} & \Lambda^{21} \\ \Lambda^{12} & \Lambda^{22} \end{bmatrix} \begin{bmatrix} F & 0 \\ 0 & F \end{bmatrix} \begin{bmatrix} X^{1} \\ X^{2} \end{bmatrix}_{2MXM} + \begin{bmatrix} N^{1} \\ N^{2} \end{bmatrix} = >$$

$$\begin{bmatrix} FY_{circ}^{1} \\ FY_{circ}^{2} \end{bmatrix} = \begin{bmatrix} \Lambda^{11} & \Lambda^{21} \\ \Lambda^{12} & \Lambda^{22} \end{bmatrix} \begin{bmatrix} F & 0 \\ 0 & F \end{bmatrix} \begin{bmatrix} X^{1} \\ X^{2} \end{bmatrix}_{2MXM} + \begin{bmatrix} N^{1} \\ N^{2} \end{bmatrix}$$
Equation (12)

The above equation now looks similar to that for a 2X2 Multi-Input Multi-Output (MIMO) system.

In order to better understand the above equalizer, the individual equations for the different frequency terms are written down. Let $y_{circ}^1(f)$ be the value of FY_{circ}^1 at frequency f. Similarly, for other terms; we then get at frequency f:

$$\begin{bmatrix} y_{circ}^{1}(f) \\ y_{circ}^{1}(f) \end{bmatrix} = \begin{bmatrix} \Lambda^{11}(f) & \Lambda^{21}(f) \\ \Lambda^{12}(f) & \Lambda^{22}(f) \end{bmatrix} \begin{bmatrix} x^{1}(f) \\ x^{2}(f) \end{bmatrix} + \begin{bmatrix} n^{1}(f) \\ n^{2}(f) \end{bmatrix}.$$

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Now letting
$$\Delta = \begin{bmatrix} \Lambda^{11}(f) & \Lambda^{21}(f) \\ \Lambda^{12}(f) & \Lambda^{22}(f) \end{bmatrix}$$
 we get:

$$\begin{bmatrix} \hat{x}_{mmse}^{1}(f) \\ \hat{x}_{mmse}^{2}(f) \end{bmatrix}_{2X1} = \left(\Delta^{H} \Delta + \sigma^{2} I_{2X2} \right)_{2X2}^{-1} \Delta^{H} \begin{bmatrix} y_{circ}^{1}(f) \\ y_{circ}^{2}(f) \end{bmatrix}_{2X1}$$
Equation (13)

15 [0027] Once the estimates of the transmitted signal are done, at all the frequencies we obtain the

 $\begin{bmatrix} \hat{X}_{mmse}^1 \\ \hat{X}_{mmse}^2 \end{bmatrix}$ by taking the inverse Fourier transform of all the stacked signals from antennas 1, 2 at all

the frequencies from equation (13). The receiver block diagram is now similar to that shown in FIG. 4. It may be of concern that for the 2X2 case a real 2X2 matrix inversion is needed in equation (13). However, note that for the 2X2 case, there are only a total of 8 FFT's/IFFT's needed,2 FFT's of received data, 4 FFT's of channel and 2 IFFT's. As compared to the 1X1 antenna case where 3 FFT/IFFT's are needed. Since the 1X1 FDE complexity is less than or comparable to NLMS, it is expected that despite the 2X2 matrix inversion needed in equation (13), the complexity for 2X2 FDE will be less than or comparable to a 2X2 NLMS equalizer. The complexity of NLMS is linear in

the number of transmit and receive antennas. The complexity estimates for the FDE for the 2X2 antenna case is given in Table 2 below, which provides a comparison of complexity per chip for the present invention FDE approach and an NLMS based equalization approach

TABLE 2

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	M = 32, P = 16	Complexity for NLMS
FFT(Y)	2*80 X and +	
FFT(h) with zero padding	4*80 X and +	
Z = FFT(Y)*FFT(h)	4*32 X	
$z1 = \left(\left[\Delta^H \Delta + \sigma^2 I \right] \right)^{-1} Z$	4*64+4*32+4*32 X, 4*64 +	
IFFT(Z)	2*80 X and +	
Total	1120 complex X,	
	736complex +	
Operations/chip = Total/P	70 complex X,	128 complex X, 120
(2X2 case)	46 complex +	complex adds

As can be seen from Table 2, the complexity of the proposed 2X2 FDE approach is less than the 2X2 NLMS equalizer.

[0028] In FIG. 6 there is shown a block diagram of a FDE for a one transmit/one receive antenna system (1 X 1). Thirty-two samples of received chips $y = \{y_1, y_2, ..., y_{32}\}$ are sent to a length 32 FFT 604 that provides a 32 sample output. A channel response at chip level, $h = \{h_1, h_2,, h_{32}\}$ 608 is provided to a length 32 FFT whose 32 sample output is element wise multiplied with the output of FFT 604. Block 619 is disabled if the FDE is to behave like a rake receiver, if not, a multipler 609 multiplies the output of block 607 with the 32 samples from multiplier 606. An inverse FFT (IFFT) is performed by block 612, and a length 16 despreading is performed by block 614. The 16 sample output of the despreader 614 goes to a slicer or other similar circuitry.

[0029] Referring now to FIG. 7, there is shown the data 706 used at the input to despreader 614, the data along the ends 702 and 704 is discarded at the output of the IFFT 612 and is not sent to the despeader 614.

[0030] In FIG. 8 there is shown a block diagram of a FDE for a one transmit/two receive antenna system (1 X 2). The receiver depicted in FIG. 8 is substantially similar to that depicted in FIG. 6, and the signal received by each of the two antennas is processed similarly to that described

with reference to FIG. 6, with the provision that the output of a first multiplier 816 and a second multiplier 818 is summed by a summing component 811.

[0031] In FIG. 9 there is shown a block diagram of a FDE for a two transmit/one receive antenna system (2 X 1). The receiver depicted in FIG. 9 is substantially similar to that depicted in FIG. 6, and the signal received by the one antenna is processed similarly to that described with reference to FIG. 6, with the provision that the input y is processed by similar receiver stages each provisioned with an appropriate channel response.

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[0032] In FIG. 10 there is shown a block diagram of a FDE for a two transmit/two receive antenna system (2 X 2). The receiver depicted in FIG. 8 is substantially similar to that depicted in FIG. 6 and combines one or more aspects of the processing depicted in FIG. 8 and FIG. 9.

[0033] In Fig. 11 there is shown a block diagram of a frequency domain equalizer architecture in accordance with another embodiment.

[0034] While the preferred embodiments of the invention have been illustrated and described, it will be clear that the invention is not so limited. Numerous modifications, changes, variations, substitutions and equivalents will occur to those skilled in the art without departing from the spirit and scope of the present invention as defined by the appended claims.

CLAIMS

- 1. A method for performing frequency domain equalization on data, comprising:
- (a) receiving data that needs to be demodulated, the data including a first vector Y_1 and a first channel impulse response, H_1 ;
- (b) performing Fast Fourier Transforms (FFT) on the first vector Y₁ and on the first channel impulse response H₁;
- (c) multiplying the data by a term of the FFTs of the first vector Y_1 and the first channel impulse response H_1 ; and
 - (d) performing an Inverse FFT (IFFT) on the result of (c).
- 2. The method of claim 1, further comprising:
 - (e) truncating portions of data resulting from (d) in order to get chip level estimates.
- 3. The method of claim 2, further comprising:
 - (f) despreading the truncated data in order to produce symbol level estimates.
- 4. The method of claim 1 or 3, further including:
 - (g) receiving a second channel impulse response H₂;
 - (h) performing a FFT on the second channel impulse response H₂;
- (j) multiplying the data by a term of the FFTs of the first vector Y₁ and the second channel impulse response H₂; and
 - (k) performing an IFFT on the result of (j).
- 5. The method of claim 1 or 4, wherein further received data is frequency domain equalized by selecting other extended blocks of data that overlap with the previous extended block of data.
- 6. The method of claim 1, further including:
 - (m) receiving a second vector Y₂ and a second channel impulse response H₂;
- (n) performing a FFT on the second vector Y_2 and on the second channel impulse response H_2 ;
- (o) multiplying a term of the FFT of the second vector Y_2 by a term of the second channel impulse response H_2 ; and

(p) summing the result of (o) into (c) before performing (d).

7. A radio receiver, comprising:

one or more data Fast Fourier Transformers, each data Fast Fourier Transformer operable to perform a Fast Fourier Transform on an input data block; one or more impulse response Fast Fourier Transformers, each impulse response Fast Fourier Transformer operable to perform a Fast Fourier Transform on a channel impulse response; one or more multiplier components operable to multiply a term of the output of one of the data Fast

Fourier Transformers by a term of the output of one of the impulse reponse Fast Fourier

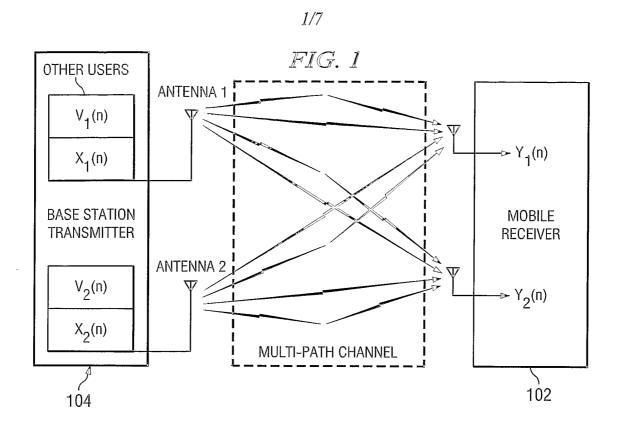
Transformers; and

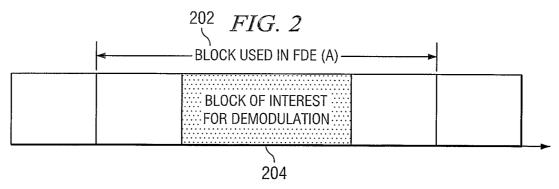
one or more Inverse Fast Fourier Transformers, each Inverse Fast Fourier Transformer operable to perform an Inverse Fast Fourier Transform based on an output of one or more of the multipliers.

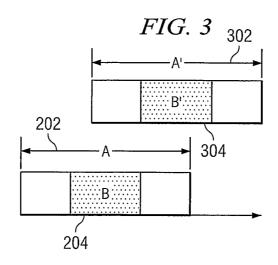
- 8. The radio receiver of claim 7, wherein two data Fast Fourier Transformers, two impulse response Fast Fourier Transformers, two multiplier components, and at least one Inverse Fast Fourier Transformer are employed.
- 9. The radio receiver of claim 8, further including a summation component operable to sum the output of the two multiplier components and provide the output to the Inverse Fast Fourier Transformer.
- 10. The radio receiver of claim 8 or 9, further comprising means for truncating the edges of the output of one or more of the multipliers to determine chip level estimates.
- 11. The radio receiver of claim 10, further comprising one or more despreaders operable to despread an output of one of the Inverse Fast Fourier Transformers to produce symbol level estimates.
- 12. A method for performing frequency domain equalization on data, comprising:
- (a) receiving a first vector Y_1 , a first channel impulse response H_{11} , and a second channel impulse response H_{21} ;

(b) receiving a second vector Y_2 , a third channel impulse response H_{12} , and a fourth channel impulse response H_{22} ;

- (c) performing Fast Fourier Transforms (FFT) on the first vector Y_1 , the second vector Y_2 , the first channel impulse response H_{11} , the second channel impulse response H_{21} , the third channel impulse response H_{12} , and the fourth impulse response H_{22} ;
- (d) determining a first product by multiplying the FFT of the first vector Y_1 by the FFT of the first channel impulse response H_{11} , a second product by multiplying the FFT of the first vector Y_1 by the FFT of the second channel impulse response H_{21} , a third product by multiplying the FFT of the second vector Y_2 by the FFT of the third channel impulse response H_{12} , and a fourth product by multiplying the FFT of the second vector Y_2 by the FFT of the fourth impulse response H_{22} ; and
 - (e) determining a first sum based on summing the first, second, third, and fourth products;
 - (f) determining a first output based on performing an Inverse FFT (IFFT) of the first sum;
- (g) determining a second sum based on summing the first, second, third, and fourth products; and
 - (h) determining a second output based on performing an IFFT of the second sum.







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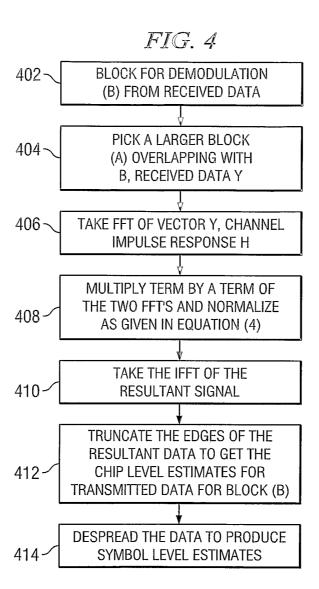


FIG. 5

