

Adaptive Interference Cancellation Using CMA-Correlation Normalized LMS for WCDMA System

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Abstract— In this article, we proposed a new interference canceller using the adaptive algorithm. We designed constant modulus algorithm-correlation normalized least mean square (CMA-CNLMS) for wireless system. This structure is normalized LMS algorithm using correlation between the desired and input signal for cancelling the interference signals in the wideband code division multiple access (WCDMA) system. We showed that the proposed algorithm could improve the Mean Square Error (MSE) performance of LMS algorithm. MATLAB (Matrix Laboratory) is employed to analyze the proposed algorithm and to compare it with the experimental results. The MSE value of the LMS with $\mu=0.0001$ was measured as -12.5 dB, and that of the proposed algorithm was -19.5 dB which showed an improvement of 7dB.

Index Terms—Interference Cancellation System (ICS), LMS algorithm, Radio Frequency (RF) repeater.

I. INTRODUCTION

In the wireless communication system, the interference signals may result in the degradation of the network performance. Adaptive filtering has been a general technique to cancel such interference signals in the communication network. Wideband Code Division Multiple Access (WCDMA) using the spread-spectrum signaling has gained increased interests for application in communication systems. Advantages of the WCDMA are asynchronous operation, a better channel usage compared with other techniques that allow a single user to be transmitted over the channel at a certain time. The interference signal is caused by communication systems. When the interference signals is detected, one solution to inhibit the divergence of the tap coefficients can be obtained by setting the step size of the adopted adaptive algorithm to zero [1].

In this paper, we propose a Constant Modulus Algorithm combined with the Correlation Normalized Least Mean Square (CMA-CNLMS) algorithm. The proposed CMA-CNLMS algorithm utilizes correlation

between the desired and input signals to adjust the step sizes, for canceling the interference signals in the WCDMA system. The proposed algorithm commonly provides better Mean Square Error (MSE) performance than the conventional LMS algorithm.

Utilizing an efficient method to calculate the tap input power required in the NLMS algorithm, a small increase of the computational complexity relative to the conventional LMS algorithm is required for updating the filter coefficients in adaptive algorithm [2]. Ref. [3] used several step sizes for an algorithm. The proposed detector uses the CMA-CNLMS algorithm with two different step sizes. The proposed algorithm is more flexible and more adaptive in multipath environment. In this structure, errors which are obtained from each group are compared, and a minimum error is chosen to the selection block. The structure has a fast convergence and mean square error performance improvement simultaneously.

The rest of this paper is organized as follows. We present basic structure for the proposed adaptive algorithm in chapter 2. Simulation and measurement results will be in chapter 3 and conclusion follows in chapter 4.

II. PROPOSED ALGORITHM

We propose a CMA-CNLMS algorithm. The cost function of CMA is a measure of the derivation of the modulus of the output $y(n)$ according to Eq.(1).

$$J_{pq} = E \left[\left| \|y(n)\|^p - 1 \right|^q \right] \quad (1)$$

Where the filter output $y(n)$ results from the weight vector $W(n)$ operating on the input signal $x(n)$, i.e.,

$$y(n) = W^T(n)x(n) \quad (2)$$

For practical reasons, p and q are positive integers and limited to 1 and 2. Then the corresponding algorithm has yielded useful adaptive form and they are referred to as CMA_{p-q} algorithm. To minimize the cost function over the space of weight vectors, a stochastic approximation to the steepest descent algorithm is used

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Weight vectors are updated as Eq.(3).

$$W(n+1) = W(n) - \mu \nabla J_{pq}(n) \quad (3)$$

Where μ is a small real scalar named step size and $\nabla J_{pq}(n)$ is an unbiased estimate of the gradient of the cost function with respect to W .

Weight vectors are updated approximately as Eq.(4) [4].

$$W(n+1) = W(n) - 2\mu e(n)x(n) \quad (4)$$

Where $e(n)$ is error signal given as Eq.(5).

The difference between the desired signal $d(n)$ and the filter output $y(n)$ is the error signal given by Eq.(5).

$$e(n) = d(n) - y(n) = d(n) - W^T(n)x(n) \quad (5)$$

In the LMS algorithm, the MSE $E[e^2(n)]$ is minimized by continuously updating the weight vector $W(n)$ as each new data sample is received, according to the Eq.(6).

$$W(n+1) = W(n) + \psi e(n)x(n) \quad (6)$$

Where ψ is a positive constant that governs the rate of convergence. The NLMS algorithm is realized by changing the ψ as Eq.(7).

$$\psi = \frac{\alpha}{x^T(n)x(n) + \beta}, 0 < \alpha < 2 \quad (7)$$

Where α is a positive constant used to ensure stable convergence, β is a positive constant used to avoid that the denominator results in zero values [2].

In this paper, the new structure proposed is based on the processing of autocorrelation function of the input signal and the cross correlation of the input signal and desired signal. The autocorrelation function $R_{xx}(n, k)$ for the input signal, with time-lag k , is defined as Eq.(8).

$$R_{xx}(n, k) = \sum_{j=0}^n x(j)x(j-k) \quad (8)$$

Also the cross-correlation between the desired and input signals is calculated as Eq.(9).

$$R_{dx}(n, k) = \sum_{j=0}^n d(j)x(j-k) \quad (9)$$

Where $x(n)$ is input signal of the adaptive filter and $d(n)$ is the desired signal.

Assuming that there is no correlation between the far-end signal $s(n)$ and the near-end signal $u(n)$, i.e. $R_{su}(n, k) \approx 0$, the cross-correlation will be obtained as follows.

$$R_{dx}(n, k) = \sum_{i=0}^n r_i R_{xx}(n, k-i) \quad (10)$$

Where r_i is the impulse response.

We need to process the autocorrelation values of the input by an adaptive filter to estimate $R_{dx}(n, 0)$.

$$\tilde{R}_{dx}(n, 0) = \sum_{i=0}^n W_i(n) R_{xx}(n, i) \quad (11)$$

Where $\tilde{R}_{dx}(n, 0)$ is the estimation of the cross-correlation for time-lag $k=0$. $R_{dx}(n, 0)$ is estimated by adjusting weight vector W_i in the same manner as the LMS algorithm.

The estimation error $g(n)$ for the correlation function is defined as follows.

$$g(n) = R_{dx}(n) - \tilde{R}_{dx}(n) \quad (12)$$

We used the minimum of the MSE for tap adaptation in the correlation adaptive filter. MSE of the correlation function is given as Eq.(13).

$$MSE = E[|R_{dx}(n) - \tilde{R}_{dx}(n)|^2] \quad (13)$$

The gradient of the MSE is

$$\nabla_j [MSE] = -2E[g(n)R_{xx}(n, j)], 0 \leq j \leq N-1 \quad (14)$$

Weight vector is updated as Eq.(15) in NLMS algorithm.

$$W(n+1) = W(n) + \frac{2\mu_0}{1 + x^T(n)x(n)} e(n)x(n) \quad (15)$$

Where μ_0 is the step size for tap coefficients adaptation [5]. In the NLMS algorithm[6,7], normalization is used to ensure sufficient conditions for convergence.

Fig. 1 shows the proposed algorithm. Errors which are obtained from each group are compared in selection part, and a minimum error is chosen. In Fig. 1, error signal $e_1(n)$ is given as Eq.(16), and $e_2(n)$ is given as Eq.(17).

$$e_1(n) = d(n) - y_1(n) \quad (16)$$

$$e_2(n) = (y_2^2(n) - R_c^2(n)) \cdot y_2(n) \quad (17)$$

In the proposed CMA-CNLS algorithm, filter coefficients for each group are updated by using output information of the selection block. In the proposed CMA-CNLS algorithm, step size is used by group. So this algorithm is more flexible and adapt itself accordingly to retain the desirable performance characteristics of fast convergence and low steady-state error.

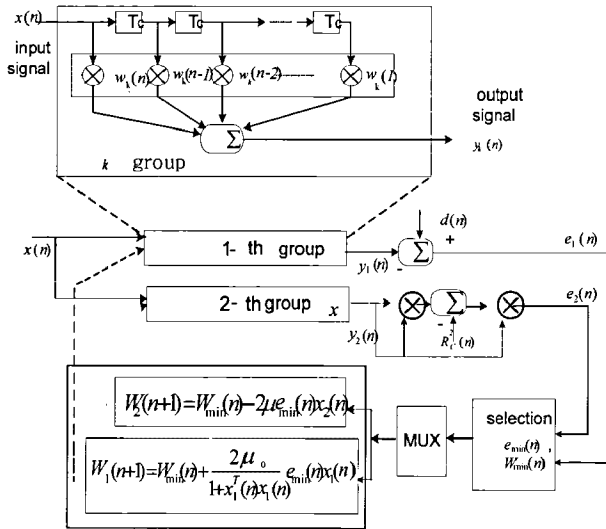


Fig. 1. Structure of the proposed interference canceller.

III. SIMULATION AND MEASUREMENT RESULTS

Fig. 2 shows the simulation results for comparing the performance of the proposed CMA-CNLS algorithm with LMS algorithm. From the Fig. 2, we can see that convergence times for the about the same performance are about the same for two algorithms. However, the proposed algorithm has better steady state MSE performances by 7 to 13.5 dB than LMS algorithm.

Fig. 3 shows the comparison for mean square error vs. iteration time for NLMS, Normalized Constant Modulus Algorithm (NCMA), Variable Step Size LMS (VSSLMS) algorithms and the proposed algorithm. The proposed algorithm has a little longer iteration time than other algorithms, but the proposed algorithm has better

steady state MSE performances by 2.5 to 10 dB than other algorithms.

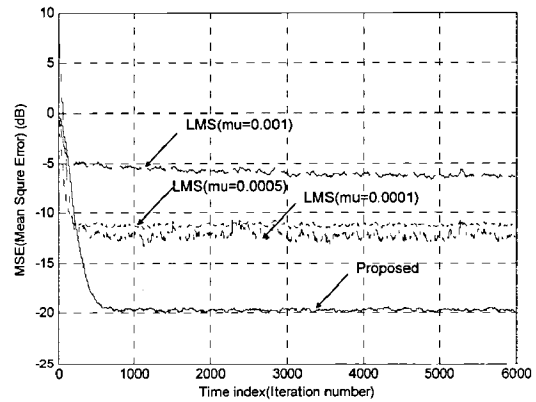


Fig. 2. Mean square error and convergence performance of the proposed algorithm and LMS algorithm with different step sizes.

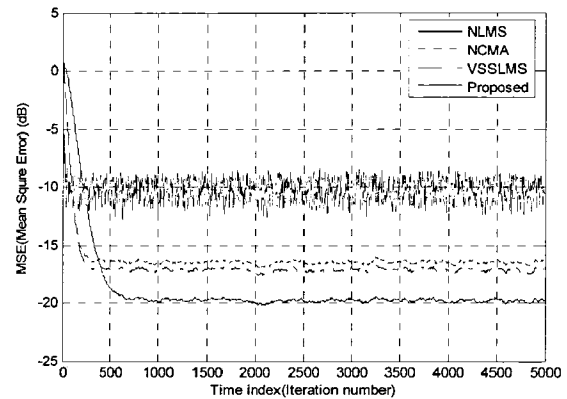


Fig. 3. Mean square error and convergence performance of the proposed algorithm and other algorithms.

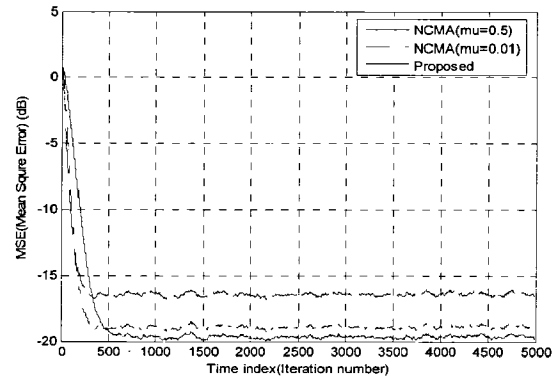


Fig. 4. Mean square error and convergence performance of the proposed algorithm and NCMA algorithm with different step sizes.

Fig. 4 shows the comparison for mean square error vs. iteration number between NCMA with constant step sizes

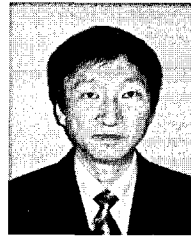
and the proposed algorithm. Also, the proposed algorithm has better MSE performances than NCMA.

IV. CONCLUSIONS

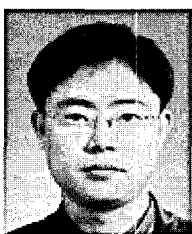
We proposed the CMA-CNLMS algorithm which utilized the correlation between the desired and input signal to adjust the step sizes. Error sequence is utilized to accelerate the speed of convergence. And, we investigated and compared the performance of the proposed CMA-CNLMS algorithm with conventional LMS, NLMS, NCMA, VSSLMS. The results showed that the proposed algorithm had a better performance in view point of steady state MSE. Comparing with conventional LMS algorithm, the proposed one showed about the same convergence time for the about same MSE performance and showed a better steady state MSE performance. The proposed algorithm should be useful for wireless system such as WCDMA systems.

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