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A time domain anti-jamming processing algorithm for GNSS ranging receiver

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Traditional anti-jamming algorithms lead to a deviation of the Global Navigation Satellite System receiver's ranging value, which has become the main obstacle to the improvement of the accuracy of the high-precision ranging receiver. Traditional time-domain anti-jamming (TDAJ) algorithms can cause signal distortion, resulting in ranging bias. This study aims to solve this problem by proposing a design method for adding preprocessing filters in time-domain anti-jamming navigation receivers. In this method, an optimal order filter for adaptively correcting signal distortion is designed according to the time-domain anti-jamming filter coefficients. The experimental results show that the algorithm proposed in this paper can effectively solve the problem of ranging bias caused by the traditional time-domain anti-jamming algorithm. The algorithm can be widely used in dedicated satellite missions that require high positioning accuracy, such as navigation signal monitoring receivers, satellite payload receivers, and precision approach and landing systems for carrier-based aircraft.

KEYWORDS

global navigation satellite system (GNSS), ranging receiver, time-domain, antijamming, ranging bias, distortion correction

1 Introduction

The electromagnetic environment of Global Navigation Satellite System (GNSS) has become increasingly complex [1–5]. According to spectrum management and planning, the main frequency band for satellite navigation is in the L-band, but there are also systems such as communication and radar in this frequency band [6]. Therefore, satellite navigation may be subject to unintentional interference [7]. According to the literature [8, 9], the most common type of interference that has the greatest impact on GNSS receivers is narrow-band interference. GNSS navigation receivers using antennae array and single antenna processing have become an important means of anti-jamming [10]. At the same time, because there is no problem of large pseudo-range deviation caused by the mismatch of the amplitude and phase characteristics of the array antenna [11, 12], single-antenna anti-jamming technology exerts its unique advantages in some scenes that require high-precision positioning [13, 14]. High-precision positioning scenarios include navigation

signal monitoring receivers, satellite payload receivers, carrier-based aircraft precision approach and landing systems, *etc.*

Compared with the communication system, the navigation system pays more attention to the ranging performance [15–17]. Thus, in the satellite navigation system, it is necessary to effectively suppress interference while reducing the impact of interference suppression technology on the system ranging performance as much as possible [18, 19]. It has been proved that under the condition of ideal receiver channel characteristics, whether it is a coherent phase detector or a non-coherent phase detector, the code phase tracking loop phase detection result of the navigation signal after the TDAJ filter is unbiased. Thus the TDAJ filter will not bias the pseudo-range measurement results [20].

Unfortunately, channel characteristics are non-ideal in actual situations, and channel non-ideal characteristics are mainly reflected in phase nonlinearity and amplitude non-flatness [21]. The existing anti-jamming filter will cause the receiver pseudo-range measurement bias under the channel non-ideal characteristics [22, 23], and the bias is related to the interference parameters [24, 25]. The bias of the ranging value caused by the interference module has become the main obstacle for the high-precision ranging receiver to improve its accuracy. Therefore, in order to achieve high-precision measurement, researchers have proposed many methods to reduce deviation. A method is to eliminate the non-ideal factors of the receiving channel by adding a calibration filter behind the analog channel. This technology is divided into analog domain equalization technology and digital domain calibration technology [26–28]. The equalization technology in the analog domain generally uses analog components such as inductors and capacitors to build an equalization network, which has the disadvantages of poor equalization accuracy and poor applicability [29, 30]. Calibration techniques in the digital domain usually have high computational complexity and poor precision control. Another type of method uses a small loop self-calibration method to correct the pseudo-range measurement of the receiver [31]. However, the coupling effect of the characteristics of the transmitting channel and the characteristics of the receiving channel may cause the delay correction value to deviate, and this method will add a transmitting channel dedicated to calibration and a receiving channel dedicated to calibration, greatly increasing the hardware complexity.

Although a great deal of work has been done to study the calibration elimination of deviations, it is still challenging to achieve good anti-jamming performance and high precision ranging [32–37]. In order to solve the above problems, this paper theoretically analyzes the influence of the TDAJ filter on the high-precision measurement of the navigation signal pseudo-range. Because the symmetry of the correlation peak is destroyed when the channel characteristics are not ideal, and the anti-jamming process of the TDAJ filter is the process of

accumulation of the correlation function. So the asymmetric correlation peak through the anti-jamming filter will cause more serious distortion. Aiming at the problem of signal distortion caused by the anti-jamming filter, this paper designs an adaptive correction filter according to the filter coefficient of the navigation signal after passing through the anti-jamming module, and preprocesses the local pseudocode signal. The algorithm effectively solves the problem of ranging bias caused by the traditional TDAJ filter.

The rest of this paper is organized as follows. First, the non-ideal channel model and signal model are established in Section 2. Second, a detailed theoretical analysis in Section 3 describes the ranging bias caused by the anti-jamming filter. Then an adaptive correction filter is designed in Section 4, and the optimal order of the filter is theoretically derived. At the same time, Section 4 also analyzes the effect of the algorithm on the carrier-to-noise ratio. Section 5 presents the simulation results of ranging bias compared with traditional algorithms, further demonstrating the effectiveness and robustness of the proposed method. Finally, Section 6 shows the conclusion.

2 Mathematical model

As shown in Figure 1, the mathematical model structure diagram shows that $S_{RF}(t)$ is the received radio frequency signal. The intermediate frequency signal $S_{IF}(t)$ is obtained after A/D sampling and digital down-conversion, and then gets output signal $S_{AJM}(t)$ after passing through the anti-jamming module. After capturing the tracking module, $S_{AJM}(t)$ is related to the local pseudocode, and the correlation function is obtained. Finally, the accumulated value of correlation is sent to the discriminator.

2.1 Channel characteristic model

In the simulation experiment, to make use of the measured channel characteristic data, additional channel characteristics can be disturbed according to the method which is shown in Figure 2 [38].

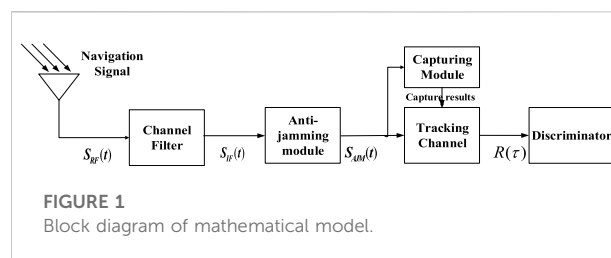
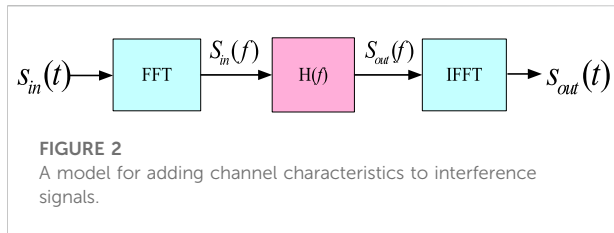


FIGURE 1
Block diagram of mathematical model.



In Figure 2, $S_{in}(t)$ is the received if signal. $S_{in}(f)$ is the output transformed to the frequency domain, $S_{out}(f)$ is the output after an equivalent nonlinear phase low-pass filter, and $S_{out}(t)$ is the output transformed to the time domain by inverse Fourier transform. The output expression of the signal after passing through the low-pass filter is:

$$S_{out}(f) = S_{in}(f)H(f) \quad (1)$$

In engineering applications, the amplitude-phase characteristics of antennas and RF channels can be measured by signal sources and spectroradiometers. Figure 3 shows the measured amplitude-phase characteristics of four single antenna channels measured by signal source and spectrum analyzer. It can be seen from Figure 3A that the magnitude response of the actual channel is non-flat, while it can be seen from Figure 3B that the phase response of the actual channel is nonlinear. The amplitude-phase characteristics of the channel in the simulation experiment will be simulated by using the actual measured data.

2.2 Signal model

The time-domain adaptive anti-jamming technique based on interference estimation is to make use of the difference in predictability between narrowband interference and wideband signal, so as to form the estimated value of narrowband interference, and then cancel it from the received signal to eliminate the narrowband interference.

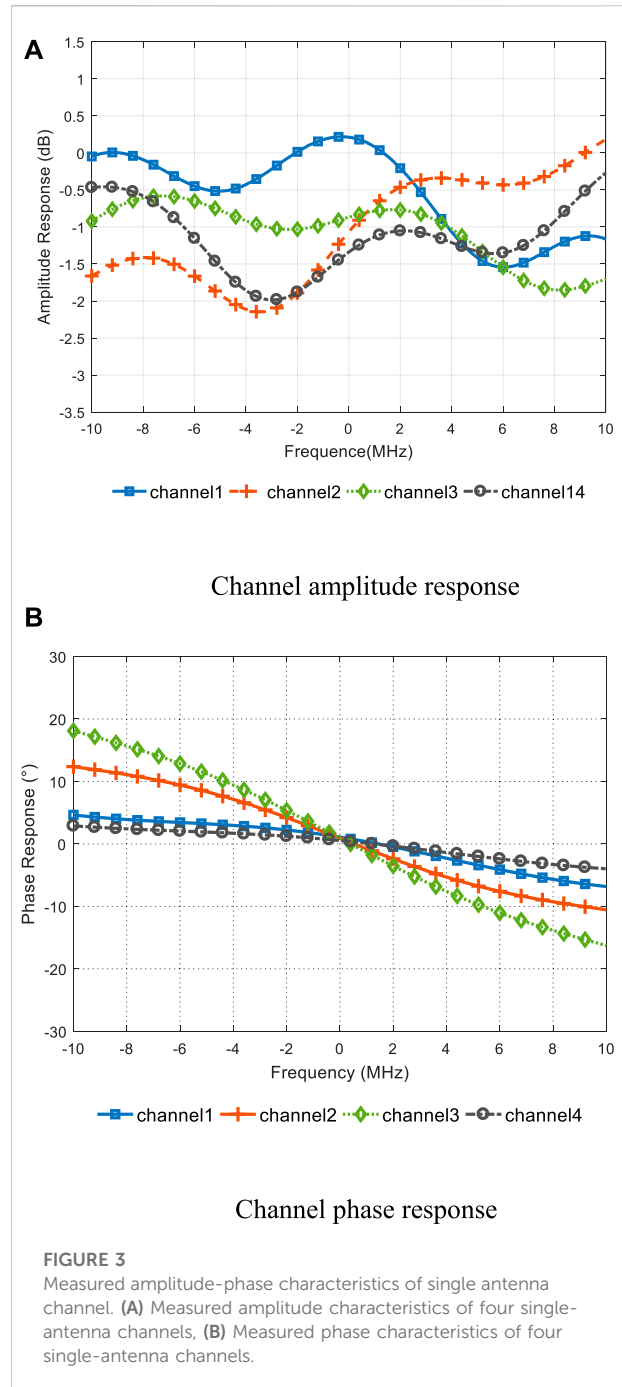
According to the literature [31], the correlation function after the signal passes through the non-ideal channel is as follows:

$$R(\tau) = \int_{-b}^b \text{sinc}^2(f) \cdot A(f) \cdot \exp\{j2\pi f[\tau - \tau_g(f)]\} df \quad (2)$$

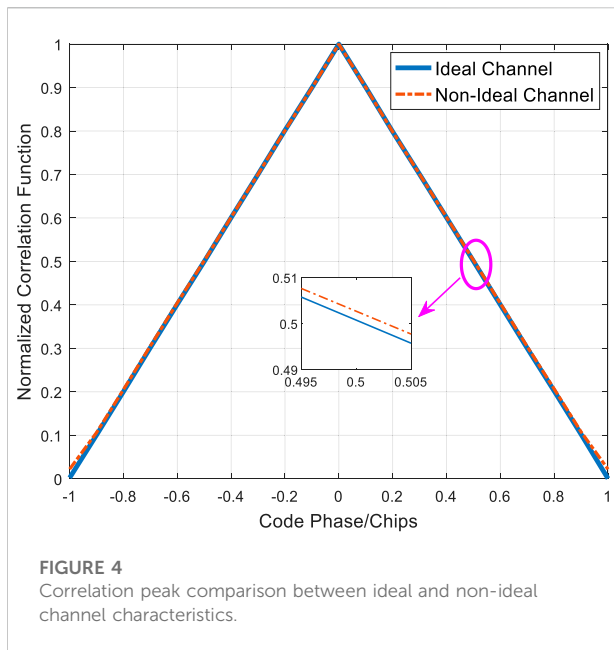
where $\tau_g(f)$ is the group delay of the equivalent low-pass filter.

It can be seen from Eq. 2 that when the navigation signal passes through the non-ideal channel, if the ideal constant amplitude and linear phase filter cannot be guaranteed, the correlation peak will no longer be symmetrical, resulting in distortion and thus affecting the ranging performance.

Figure 4 shows the correlation function when navigation signals pass through ideal channel and non-ideal channel



respectively. But from the Figure 4, we cannot intuitively distinguish the difference between the two correlation peaks. Therefore, in the satellite navigation signal channel indicator system, S-curve Bias (SCB), an important indicator, is usually used to measure the symmetry of correlation peaks and quantitatively analyze the ranging bias of receivers [39, 40]. The definition of SCB is expressed as follows, where τ_0 , τ_1 and τ_2 meet the following constraints:



$$\tau_{SCB} = \tau_0 - \frac{|\tau_1 - \tau_2|}{2}, \begin{cases} \tau_0 = \operatorname{argmax} R(\tau) \\ R(\tau_1) = R(\tau_2) \\ |\tau_1 - \tau_2| = \Delta \end{cases} \quad (3)$$

In Eq. 3, $R(\tau)$ is the correlation function between the received signal and the local signal, and Δ is the interval of the correlator.

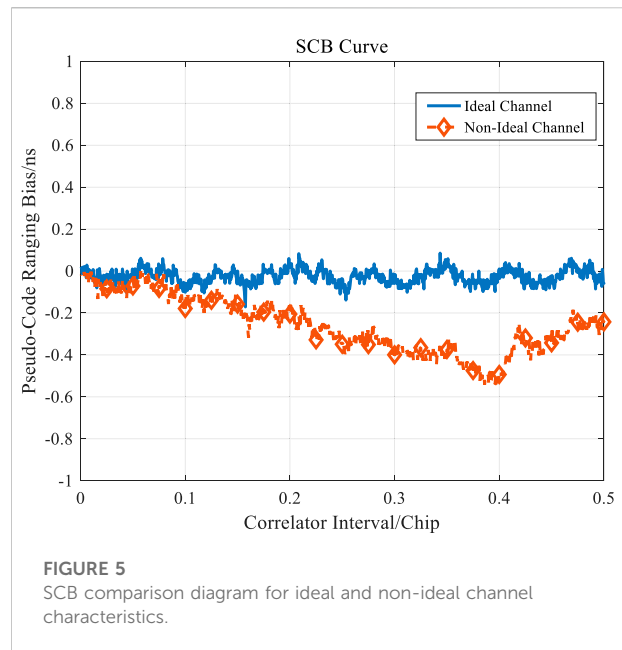
As shown in Figure 5, under ideal and non-ideal channel characteristics, the maximum pseudocode ranging deviation exceeds 0.5 ns. Once again it is proved that the error of ranging is caused by the navigation signal passing through the non-ideal channel.

2.3 TDAJ filter

In the single-antenna-based interference suppression algorithm, we adopt the time-domain interference suppression based on linear estimation with mature technical basic theory. In this paper, combined with the current situation of engineering implementation, the classical minimum mean square error (LMS) algorithm [41, 42] is selected to provide convenience for the following analysis and application.

In addition, the filter structure adopted in this paper is the common odd-order bilateral tapped transverse filter, namely the filter with interpolation structure. It uses both past and future data to estimate $x(n)$ based on the data set $\{x(0) \cdots x(n-1) \ x(n+1) \cdots x(N-1)\}$.

The order of the filter is $(2M + 1)$. In addition, to prevent the filter coefficient from converging to 0 and facilitate calculation, the intermediate weight of the filter is fixed as 1. In the subsequent studies of this paper, the odd-order bilateral



tapped lateral filter is taken as an example for analysis. For an even-order unilateral tapped filter, the local pseudocode signal can be obtained by appropriate delay.

3 Measurement performance analysis

Because the non-ideal characteristics of the receiving channel will distort the correlation peak of the received signal, the correlation peak of the signal input to the adaptive anti-jamming filter no longer meets the requirement of symmetry. In section 2, we have presented the theory and simulation verification. However, in the case of non-ideal channel characteristics, the superposition effect of correlation functions will be generated in the LMS algorithm iteration process when the navigation signals pass through the TDAJ filter, which will aggravate the distortion of correlation peaks and lead to serious bias in ranging. In the following part, the specific analysis will be carried out theoretically.

The structure of the adaptive filter and the implementation of the LMS algorithm are mentioned in Section 2.3. The input of the filter is as follows:

$$\mathbf{x} = \{x(n-M) \cdots x(n-1) x(n) x(n+1) \cdots x(n+M)\} \quad (4)$$

where, the TDAJ filter coefficient \mathbf{w} is:

$$\mathbf{w} = [w_{-M} \ w_{-M+1} \ \cdots \ w_{-1} w_0 \ w_1 \ \cdots \ w_M]^T \quad (5)$$

In addition, it has been proved in literature [25] that in the usual way of adaptive filtering, when the filter coefficient meets the conjugate symmetry property, the tracking property of the

receiver can be maintained and the SNR loss can be better met. Therefore, when studying the influence of an anti-jamming filter on ranging deviation, the coefficient of the anti-jamming filter is usually kept in conjugate symmetry, as shown in Eq. 6:

$$w_n = w_{-n}^* \quad n = 1, 2, \dots, M \quad (6)$$

Therefore, the discrete output $y(n)$ at the current moment can be expressed as:

$$y(n) = \mathbf{w}^H \mathbf{x} = \sum_{i=-M}^M w_i x(n+i) \quad (7)$$

The output signal after passing through the anti-jamming module is correlated with the local signal $s(n)$, and the discrete form of the correlation function $R(\tau)$ can be expressed as:

$$R(\tau) = \sum_{n=-\infty}^{\infty} y(n) s^*(n-\tau) \quad (8)$$

Bring the Eq. 7 into the Eq. 8:

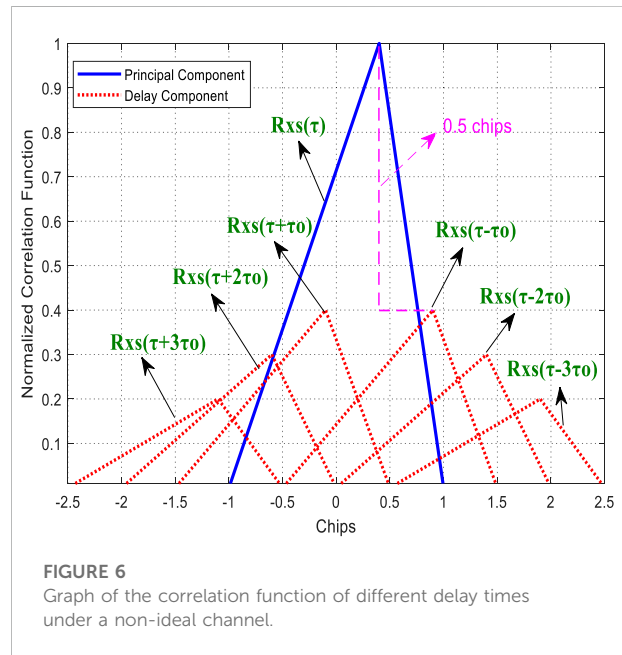
$$\begin{aligned} R(\tau) &= \sum_{n=-\infty}^{\infty} \left[\sum_{k=-M}^M w_k x(n+k) \right] s^*(n-\tau) = w_{-M} \sum_{n=-\infty}^{\infty} x(n-M) s^*(n-\tau) \\ &+ (w_{-M+1} \sum_{n=-\infty}^{\infty} x(n-M+1) s^*(n-\tau) + \dots + w_{-1} \sum_{n=-\infty}^{\infty} x(n-1) s^*(n-\tau) + w_0 \sum_{n=-\infty}^{\infty} x(n) s^*(n-\tau) \\ &+ w_1 \sum_{n=-\infty}^{\infty} x(n+1) s^*(n-\tau) + \dots + w_M \sum_{n=-\infty}^{\infty} x(n+M) s^*(n-\tau) \end{aligned} \quad (9)$$

The correlation function of the above formula can be further simplified as:

$$\begin{aligned} R(\tau) &= \sum_{k=-M}^M w_k R_{xs}(\tau + k \cdot \tau_0) = w_{-M} R_{xs}(\tau - M \cdot \tau_0) + \dots + \\ &w_{-1} R_{xs}(\tau - \tau_0) + w_0 R_{xs}(\tau) + w_1 R_{xs}(\tau + \tau_0) + w_M R_{xs}(\tau + M \cdot \tau_0) \end{aligned} \quad (10)$$

where, τ_0 is the sampling interval, and R_{xs} is the correlation function between the input signal and the local signal that has not passed the anti-jamming filter at different times. Since the interference signal, noise signal, and navigation signal are not correlated, R_{xs} is the correlation function of the navigation signal and local pseudocode at different times.

From the theoretical proof of Eq. 9 and Eq. 10, it can be seen that when the filter coefficient w_k is kept symmetric and the principal component of the correlation function R_{xs} is kept symmetric, the correlation peak after the anti-jamming filter will remain symmetric. Unfortunately, due to the non-ideal characteristics of the channel in front of the anti-jamming module, the symmetry of the correlation peak is destroyed. During the coefficient iteration process of the anti-jamming filter, M asymmetric correlation functions of lead and delay are generated, as shown in Figure 6. Because of the superposition effect of correlation functions at different times, the original asymmetric correlation peak distortion degree after



passing the anti-jamming filter is intensified, and the distortion degree is related to the anti-jamming filter coefficient.

In addition, when the anti-jamming filter coefficient converges, w_k gradually decreases to 0 on both sides based on the intermediate coefficient, so the amplitude of the correlation function at the corresponding time will also decrease correspondingly. Therefore, in general, correlation functions after the phase of τ_0 codes in advance and delay have the greatest influence on the main correlation peak, which can be well seen in Figure 6. The superposition results of the principal components of the correlation function and each delay component of the correlation function are shown in Figure 7.

It can be seen from the superposition results of correlation functions on the right of Figure 7. Due to the superposition of correlation functions at different times, the correlation summit after passing through the anti-jamming module under the characteristics of a non-ideal channel will produce more serious distortion, which will affect the range measurement deviation and the SNR, and the real value of filter coefficient convergence obtained by the LMS algorithm in the experiment was used for simulation, and the cumulative correlation peak was obtained as shown in the left of Figure 7.

4 Robust adaptive correction ranging algorithm

Aiming at the problem of pseudo-distance deviation caused by the anti-jamming module discussed above, this section will propose a robust adaptive correction algorithm, and theoretically deduce the optimal filter order under this algorithm. The impact of the algorithm will also be analyzed.

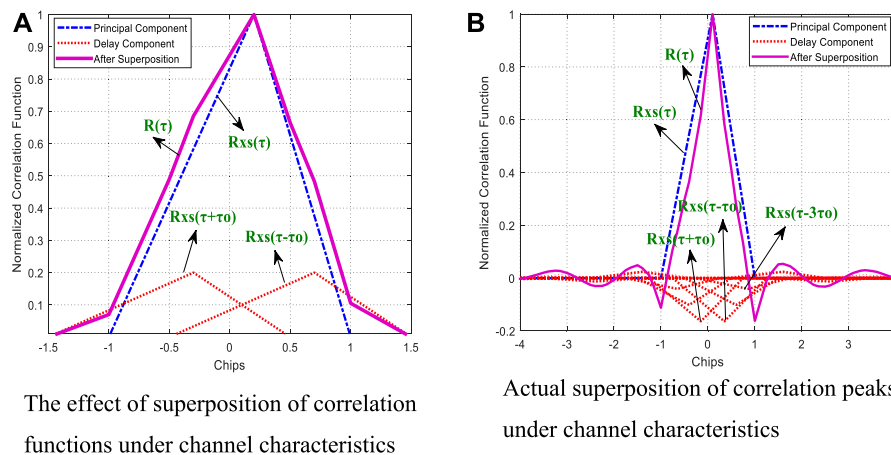


FIGURE 7

The effect of superposition of correlation functions under channel characteristics. (A) A rendering of the superposition of correlation functions under channel characteristics, (B) Actual superposition of correlation functions in traditional LMS algorithm under channel characteristics.

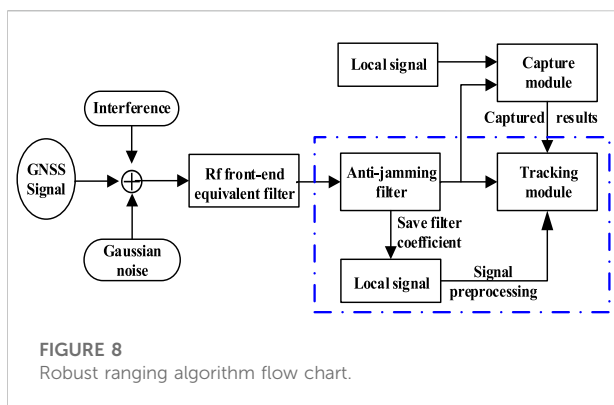
4.1 Algorithm principle

The flow chart of the robust ranging algorithm is shown in Figure 8.

First of all, in order to eliminate the superposition effect of related functions brought by the anti-jamming module in the iterative process, we will design the filter coefficient for preprocessing the local pseudocode according to the anti-jamming filter coefficient, and the specific calculation formula will be given later. The signal output y_s through the preprocessing filter is expressed as follows:

$$y_s(n) = \alpha^H s = \sum_{i=-L}^L \alpha_i s(n+i) \quad (L \leq M) \quad (11)$$

Then, the navigation signal passing through the anti-jamming filter is correlated with the preprocessed local signal, and its correlation function is expressed as follows:

FIGURE 8
Robust ranging algorithm flow chart.

$$R'(\tau) = \sum_{n=-M}^M y(n) y_s^*(\tau - n) \quad (12)$$

Substituting Eqs 7, 11 into Eq. 12 can be calculated as follows:

$$R'(\tau) = \sum_{n=1}^N \left[\sum_{k=-M}^M w_k x(n+k) \right] \left[\sum_{i=-L}^L \alpha_i s(\tau - n - i) \right]^* \quad (13)$$

Further calculation of the above related functions can be obtained as follows:

$$R'(\tau) = \left(\sum_{i=2M-L}^M w_i \alpha_{i-M-L} \right) R_{xs}(\tau + (M+L)\tau_0) + \left(\sum_{i=2M-L-1}^M w_i \alpha_{i-M-L-1} \right) R_{xs}(\tau + (M+L-1)\tau_0) + \dots + \left(\sum_{i=M-2L+1}^M w_i \alpha_{i+1} \right) R_{xs}(\tau + \tau_0) + \left(\sum_{i=-L}^L w_i \alpha_i \right) R_{xs}(\tau) + \left(\sum_{i=-M}^{2L-M-1} w_i \alpha_{i-1} \right) R_{xs}(\tau - \tau_0) + \dots + \left(\sum_{i=-M}^{L-2M+1} w_i \alpha_{i+M+1-L} \right) R_{xs}(\tau - (M+L-1)\tau_0) + \left(\sum_{i=-M}^{L-2M} w_i \alpha_{i+M+L} \right) R_{xs}(\tau - (M+L)\tau_0) \quad (14)$$

Further considering that when the order of the filter preprocessing the local pseudocode is equal to the order of the anti-jamming filter (that is, when $2L+1 = 2M+1$), the Eq. 14 can be further simplified to obtain:

$$R'(\tau) = \sum_{j=M-2L+1, k=1}^{k=M+L, j=2M-L} \left(\sum_{i=j}^M w_i \alpha_{i-k} \right) R_{xs}(\tau + k\tau_0) + \left(\sum_{i=-L}^L w_i \alpha_i \right) R_{xs}(\tau) + \sum_{j=L-2M, k=-M-L}^{j=2L-M-1, k=-1} \left(\sum_{i=-M}^j w_i \alpha_{i+k} \right) R_{xs}(\tau - k\tau_0) \quad (15)$$

When the order of the local pseudocode preprocessing filter is equal to the order of the anti-jamming filter (that is, when $2L+1 < 2M+1$), the Eq. 14 can be further simplified to obtain:

$$\begin{aligned}
R'(\tau) = & \sum_{\substack{j=M-2L+1, \\ k=M}}^{j=M} \left(\sum_{i=j}^M w_i \alpha_{i-k} \right) R_{xs}(\tau + k\tau_0) \\
& + \sum_{k=1}^{M-1} \left(\sum_{i=k-L}^{k+L} w_i \alpha_{i-k} \right) R_{xs}(\tau + k\tau_0) + \left(\sum_{i=-L}^L w_i \alpha_i \right) R_{xs}(\tau) \\
& + \sum_{k=1-M}^{-1} \left(\sum_{i=-k-L}^{L-k} w_i \alpha_{i+k} \right) R_{xs}(\tau - k\tau_0) \\
& + \sum_{\substack{j=2L-M-1, \\ k=-M-L}}^{j=2L-M-1} \left(\sum_{i=-M}^j w_i \alpha_{i+k} \right) R_{xs}(\tau - k\tau_0)
\end{aligned} \quad (16)$$

In Section 3 of this paper, it has been deduced and proved that the aggravation of correlation peak distortion is caused by the superposition effect of correlation functions at different moments after the signal passes through the anti-jamming module. Therefore, in Eq. 15 and Eq. 16, we preprocessed the local signal through the filter coefficient α_i , so that the coefficient before $R_{xs}(\tau)$ of the principal component of the correlation function is 1, and the pre-coefficient of $R_{xs}(\tau \pm k\tau_0)$ at other different delay times is 0. The equations are as follows:

$$\begin{cases} \sum_{i=-L}^L w_i \alpha_i = 1, & k = 0 \\ \sum_i w_i \alpha_{i \pm k} = 0, & k \neq 0 \end{cases} \quad (17)$$

In the form of matrix, the solution to α_i can be expressed as:

$$\begin{bmatrix} w_{-2L} & w_{-2L+1} & \cdots & w_{-1} & w_0 \\ w_{-2L+1} & w_{-2L+2} & \cdots & w_0 & w_1 \\ \vdots & \vdots & \ddots & \vdots & \vdots \\ w_{-1} & w_0 & \cdots & w_{2L-2} & w_{2L-1} \\ w_0 & w_1 & \cdots & w_{2L-1} & w_{2L} \end{bmatrix} \begin{bmatrix} \alpha_{-L} \\ \alpha_{-L+1} \\ \vdots \\ \alpha_{L-1} \\ \alpha_L \end{bmatrix} = \begin{bmatrix} 0 \\ \vdots \\ 1 \\ \vdots \\ 0 \end{bmatrix} \quad (18)$$

According to Eq. 16, it not only ensures that the principal component $R_{xs}(\tau)$ of the correlation function $R'(\tau)$ will not change, but also eliminates the superposition effect of the correlation function caused by iteration due to the anti-jamming process, and ensures robust ranging of correlation peak in the capture and tracking stage. At this time, the peak value of the correlation function closest to the peak value of $R_{xs}(\tau)$ should be two code chips, and the preprocessing filter order is $2L+1$. The formula of L is as follows, where $\lfloor \cdot \rfloor$ stands for rounding down.

$$L = \left\lfloor 2 \frac{F_s}{F_c} \right\rfloor \quad (19)$$

4.2 Optimal preprocessing filter order

In the navigation receiver, Early minus Later (E-L) code phase discriminator is used for ranging [28, 29]. In Eq. 19, if the delay component of the correlation function only affects the principal component of the correlation function beyond the E-L code interval, the ranging result will still not be affected. Therefore, when the preprocessed local signals are correlated with the signals passing through the anti-jamming filter, the robust ranging performance can be maintained by eliminating other correlation function delay components within the E-L code interval of the correlation function principal component. The code interval is 2Δ , then the optimal order of the preprocessing filter through which the local signal passes is $2L_0 + 1$, where L_0 is expressed as follows:

$$L_0 = \left\lfloor (1 + \Delta) \frac{F_s}{F_c} \right\rfloor \quad (20)$$

In Figure 9, when the sampling rate is 20 MHz and the code rate is 10.23MHz, according to Eq. 20, it can be seen that the optimal preprocessing filter order is five orders. It can be seen from the figure that in scenarios with different narrowband interference and correlator intervals, when the preprocessing filter order reaches the theoretically optimal order, the ranging value reaches a stable state, and the optimal order is effective.

4.3 Analysis of the influence of adaptive correction filter on CNR

Carrier-to-noise ratio (CNR) is defined as the ratio of signal power to the noise power in a unit bandwidth, and is an important indicator for evaluating signal quality in navigation reception [43]. In order to further derive the loss limit of the CNR of the robust ranging algorithm proposed in Section 4.1, the signal energy loss due to the superposition effect of the correlation function is analyzed from the energy perspective of the correlation domain. In the correlation domain, in the absence of interference and noise, the area of the correlation function in $[-T_c, T_c]$ is the signal energy. After passing through the anti-jamming filter, the total area of the correlation function at other time delays caused by the iterative effect is the signal loss energy. Regardless of interference and noise, the energy of the navigation signal correlation function is expressed as follows:

$$E = \int_{\tau=-T_c}^{T_c} |w_0 R_{ss}(\tau)|^2 d\tau \quad (21)$$

where, T_c represents the code width.

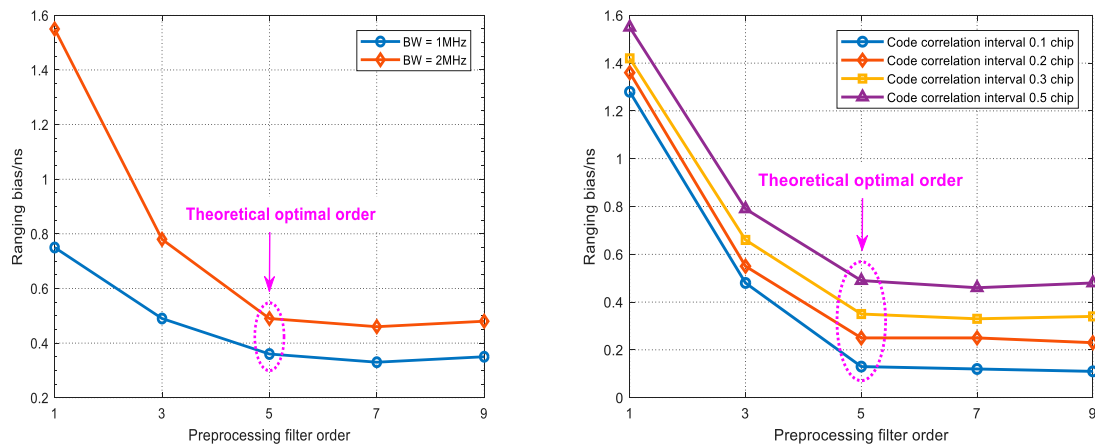


FIGURE 9
Optimal order under different interference scenarios and correlator intervals.

The total energy loss of the pure navigation signal after passing through the anti-jamming filter can be deduced as:

$$E_{Loss} = \sum_{k=-M}^M \int_{\tau=kT_0-T_c}^{kT_0+T_c} |w_k R_{ss}(\tau - kT_0)|^2 d\tau \quad (22)$$

Then the power of signal loss can be expressed as:

$$P_{Loss} = \frac{E_{Loss}/2T_c}{E/2T_c} = \frac{\sum_{k=-M}^M |w_k|^2}{|w_0|^2} \quad (23)$$

Because of the complexity of the expression of the filter coefficients in the time domain, the coefficients are derived from the ideal band-stop filter in the frequency domain. Assume that the symbols of the input signal, weight value, and output signal in frequency domain anti-jamming are $X(k)$, $H(k)$, and $Y(k)$, the corresponding symbols in the time domain are $x(n)$, $h(n)$ and $y(n)$. Since the frequency domain and the time domain satisfy the Fourier transform relationship, filtering in the frequency domain can be equivalent to filtering in the time domain. The frequency domain and time domain anti-jamming weights can be expressed as:

$$Y(k) = X(k)H(k) \quad (24)$$

$$y(n) = x(n) \otimes h(n) \quad (25)$$

Focus on analyzing the frequency domain filter $H(k)$, the relationship between $H(k)$ and $h(n)$ is as follows:

$$P_{Loss} = \frac{E_{Loss}/2T_c}{E/2T_c} = \frac{\sum_{k=-M}^M |w_k|^2}{|w_0|^2} \quad (26)$$

Assuming zero is set on the frequency interval of $[k_1, k_2]$ and the remaining interval remains 1, then the filter coefficient in Eq. 26 can be further expressed as:

$$h(n) = \text{ifft}[H(k)] = \frac{1}{N} \sum_{k=0}^{k_1-1} e^{j\frac{2\pi}{N}kn} + \frac{1}{N} \sum_{k_2+1}^{N-1} e^{j\frac{2\pi}{N}kn} = \frac{1}{N} \sum_{k=0}^{N-1} e^{j\frac{2\pi}{N}kn} - \frac{1}{N} \sum_{k_1+1}^{k_2} e^{j\frac{2\pi}{N}kn} = \begin{cases} \frac{N - (k_2 - k_1)}{N} & n = 0 \\ \frac{1}{N} \frac{e^{j\frac{2\pi}{N}k_2n} - e^{j\frac{2\pi}{N}k_1n}}{1 - e^{j\frac{2\pi}{N}n}} & n \neq 0 \end{cases} \quad (27)$$

In order to prevent the filter coefficient from converging to 0 and to facilitate calculation, the middle tap is usually constrained to one when using a bilateral tapped transverse filter. Then in the time domain, as in Eq. 27, the modulus ratio of the tapped coefficient at $n \neq 0$ to the middle coefficient is as follows:

$$w_{com_abs} = \left| \frac{1}{N} \frac{e^{j\frac{2\pi}{N}k_2n} - e^{j\frac{2\pi}{N}k_1n}}{1 - e^{j\frac{2\pi}{N}n}} \right| \quad (28)$$

Using Euler's formula to expand and further simplify:

$$w_{com_abs} = \frac{1}{N} \frac{\sqrt{\left[\cos\left(\frac{2k_2\pi}{N}\right) - \cos\left(\frac{2k_1\pi}{N}\right)\right]^2 + \left[\sin\left(\frac{2k_2\pi}{N}\right) - \sin\left(\frac{2k_1\pi}{N}\right)\right]^2}}{\sqrt{\left[1 - \cos\left(\frac{2n\pi}{N}\right)\right]^2 + \sin^2\left(\frac{2n\pi}{N}\right)}} = \frac{1}{N} \sqrt{\frac{\sin^2\left(\frac{(k_1 - k_2)\pi}{N}\right)}{\sin^2\left(\frac{n\pi}{N}\right)}} \leq \frac{1}{N} \sqrt{\frac{1}{\sin^2\left(\frac{n\pi}{N}\right)}} = \frac{1}{N} \frac{1}{\left|\sin\left(\frac{n\pi}{N}\right)\right|} \quad (29)$$

As shown in Eq. 29, w_{com_abs} is regarded as a binary function of n and N . When $N \in [2, \infty)$, $n \in [1, N-1]$, w_{com_abs} gets the maximum value at $N = 2, n = 1$. By substituting the values of n and N into Eq. 31, the modulus ratio of tap coefficient at $n \neq 0$ to the middle tap coefficient can satisfy the following relation:

$$w_{com_abs} = \left| \frac{w_{n \neq 0}}{w_0} \right| \leq \frac{1}{2} \quad (30)$$

The signal energy loss is maximum when the delay component of the correlation function is outside the interval $[-T_c, T_c]$ (equivalent to when the delay correlation function does not overlap with the principal component of the correlation function). Under the condition of channel characteristics, taking the third-order bilateral tapped transverse filter as an example, the maximum energy loss of correlation function of pure navigation signal before and after passing through the anti-jamming filter is shown in Figure 10:

In Figure 10, the red correlation function at the top is the correlation function before the pure navigation signal passed through the anti-jamming module under the channel characteristics, and the area of the correlation function within the interval $[-T_c, T_c]$ is the signal energy. Therefore, under the N-order filter coefficient, the maximum signal power loss caused by the superposition effect can be obtained as follows.

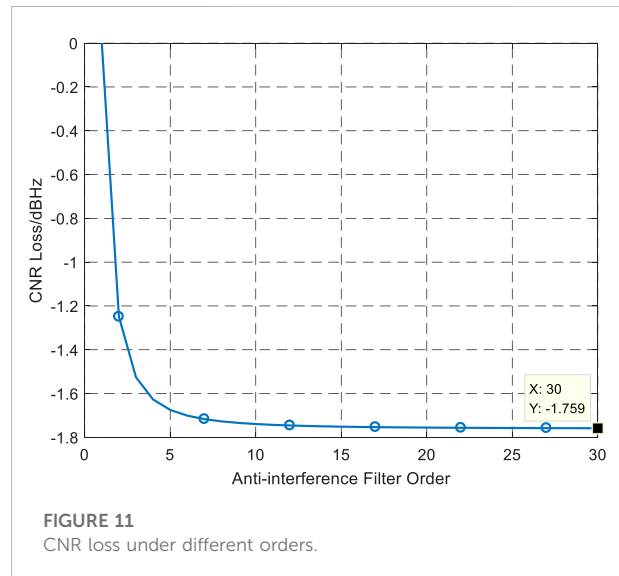
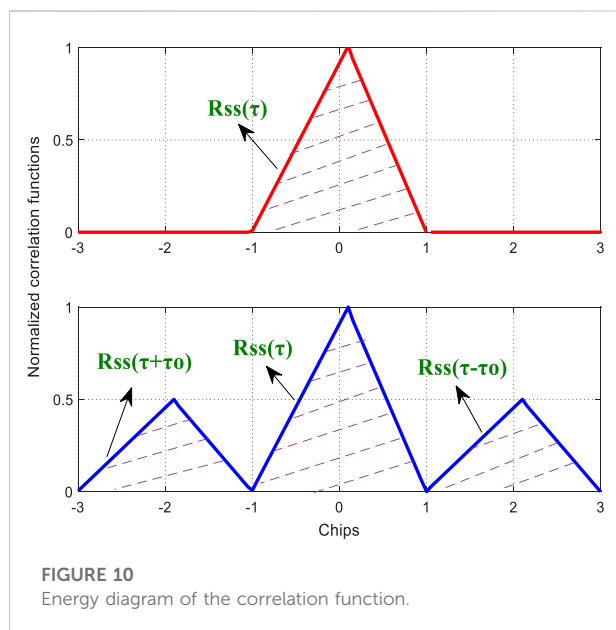
Substitute Eq. 29 into Eq. 23:

$$Loss_{CFS_MAX} = \frac{E_{Loss}/2T_c}{E/2T_c} = \sum_{n=1}^{N-1} \left[\frac{1}{N} \frac{1}{\left| \sin\left(\frac{n\pi}{N}\right) \right|} \right]^2 \quad (31)$$

By numerical simulation of Eq. 31, it can be seen from Figure 11 that when N is greater than order 10, the CNR loss caused by signal superposition tends to a fixed value of 1.8 dB (one decimal place is reserved). It is concluded that the maximum signal power loss caused by the adaptive correction algorithm is 1.8 dB.

5 Simulation verification results

In this section, the performance of a typical Beidou ranging receiver under different interference conditions will be simulated to analyze the ranging deviation caused by the anti-jamming module,



including simulation verification of the optimal order of the ranging algorithm proposed in this paper and performance analysis.

5.1 Experimental simulation parameters

5.2 Experimental simulation results

The parameters for this simulation are listed in Table 1. Where ISR represents the interference-to-signal ratio, and CNR represents the carrier-to-signal ratio, and F_s represents sampling frequency, F_c represents code rate. In addition, in the simulation,

TABLE 1 Simulation parameter setting.

Parameter settings	Value/Type
Signal type	BeiDou
F_s	20 (MHz)
Carrier frequency	1.38 (MHz)
Signal length	1,000 ms
F_c	10.23 (MHz)
Code length	10,230
CNR	57 (dB)
ISR	50 (dB)
Interference type	Narrowband interference
Anti-jamming criterion	MMSE
Anti-jamming algorithm	LMS
Filter order	29
Integral time	2 ms
Initial phase	20 (Chips)

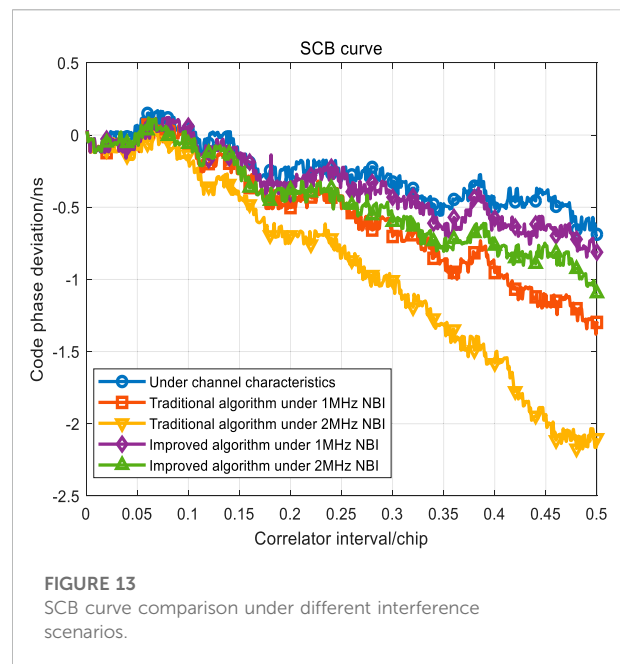
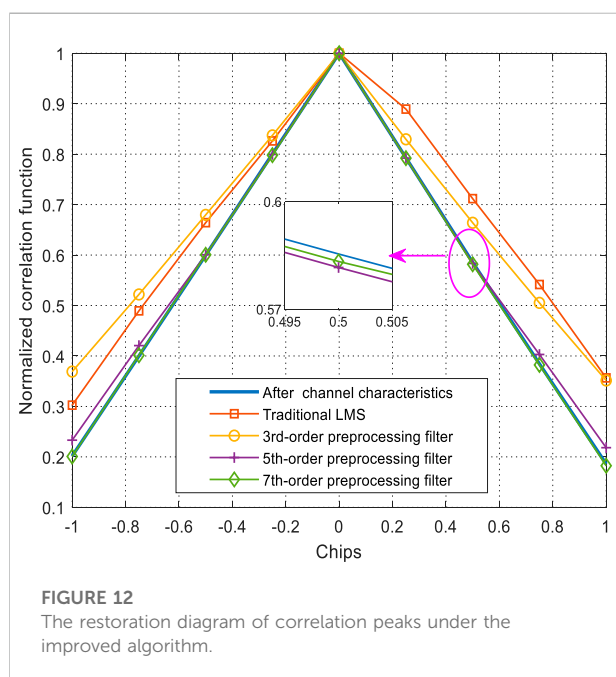
the anti-jamming filter adopts the lateral filter with a double tap. The interference types are single frequency interference and narrowband interference with different bandwidths.

It should be noted that in the following experimental scenarios, the channel characteristics adopt the measured amplitude and phase characteristics shown in Figure 3. In addition, it should be noted that in the following simulation scenarios, the ranging deviation values of the vertical coordinates are based on the ranging values under the condition of no interference.

In Figure 12, according to formula (20), it can be known that the optimal preprocessing filter order is fifth, and from Figure 14, the distorted correlation peaks after passing through the traditional LMS algorithm are all repaired to different degrees after passing through preprocessing filters of different orders. When the preprocessing filter order is greater than or equal to the optimal order, the different delay components of the correlation function cumulative in the correlator interval can be eliminated, so that the distorted correlation peak can be adaptively corrected.

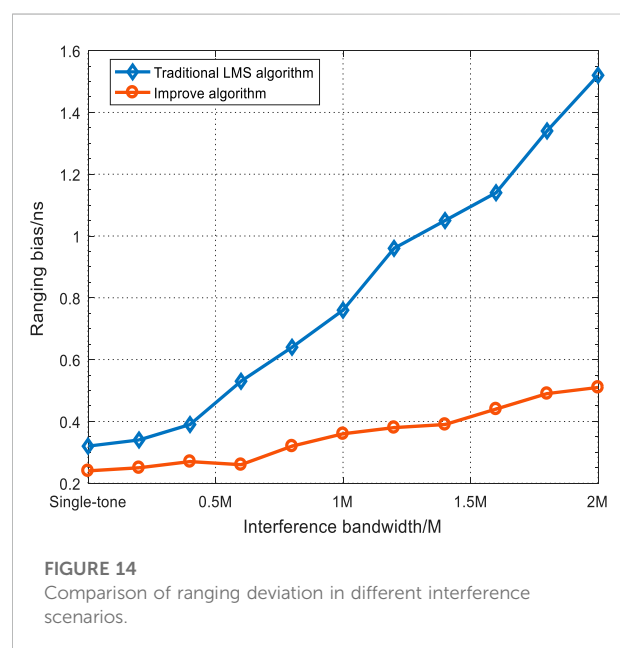
In order to better evaluate the ranging deviation at different correlator intervals, the SCB curves of the improved algorithm and the traditional LMS algorithm under different narrow-band interference are drawn. As shown in Figure 13, the blue curve is the deviation caused by the actual channel characteristics without interference. Compared with the traditional LMS algorithm, the proposed algorithm significantly reduces the ranging deviation, and the ranging value of the signal after passing through the non-ideal channel characteristics is also controlled within the index range.

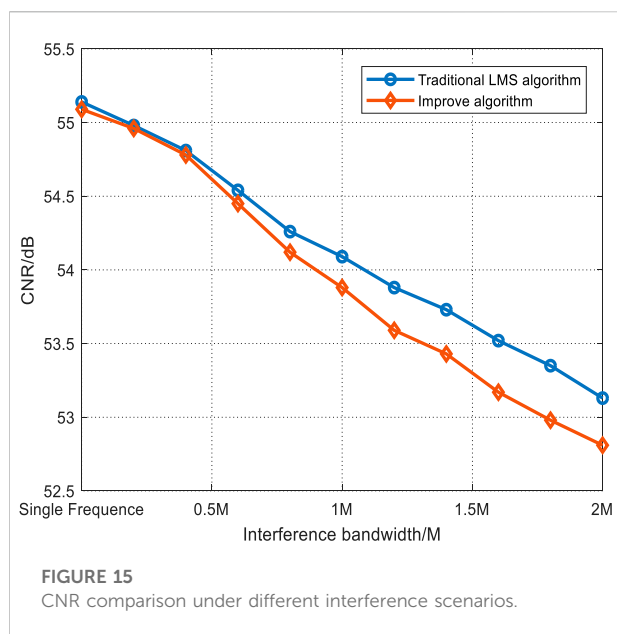
In Figure 14, compared with the ranging value of the received signal after passing through the non-ideal channel characteristic,



the ranging deviation of the traditional LMS algorithm under different narrowband interference increases with the increase of the interference bandwidth, and it can reach about 1.5ns under 2 MHz narrowband interference. The improved algorithm proposed in this paper corrects the deviation to within 0.5 ns, which conforms to the ranging performance index of the ranging receiver and achieves the effect of robust ranging.

In Figure 15, whether under the traditional LMS algorithm or the improved algorithm proposed in this paper, with the increase





of interference bandwidth, the loss of the SNR also increases. The SNR loss of the improved algorithm is slightly higher than that of the traditional LMS algorithm, and its loss is less than 1.8 dB in narrowband interference scenarios with different bandwidths, which is consistent with the theoretical analysis in section 4.3.

6 Conclusion

Due to the continuous superposition of the coefficients of the anti-jamming filter, the deviation of the measurement accuracy will be caused under the traditional LMS anti-jamming algorithm. The robust ranging algorithm proposed in this paper designs the corresponding preprocessing filter coefficients according to the anti-jamming filter coefficients. In a typical narrowband interference scenario, the ranging deviation of the traditional LMS anti-jamming algorithm is about 1.5 ns, while the algorithm proposed in this paper corrects the ranging deviation to within 0.5 ns. The simulation results show that the robust adaptive correction algorithm can adaptively correct the signal distortion and effectively reduce the measurement deviation under the optimal preprocessing filter order. At the same time, it can also maintain good anti-jamming performance.

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Data availability statement

The original contributions presented in the study are included in the article/supplementary material, further inquiries can be directed to the corresponding authors.

Author contributions

XL performed the theoretical study, conducted the experiments, processed the data and wrote the manuscript. ZL designed the system, provided research suggestions and revised the manuscript together with FC, ZL, and YM helped in performing the experiments. GO provided the experiment equipment and suggestions for the manuscript. All authors have read and agreed to the published version of the manuscript.

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Conflict of interest

The authors declare that the research was conducted in the absence of any commercial or financial relationships that could be construed as a potential conflict of interest.

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