

# Performance Enhancement of Multi-Band OFDM using Spectrum Equalizer

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**Abstract**— In this paper, equalization for frequency slope of path loss in Multi-Band(MB) OFDM UWB is proposed. The path loss of a signal is proportionate to the square of the signal's frequency. So, the received signal amplitudes of OFDM subcarrier can be different up to 3dB when MB-OFDM occupies bandwidth over 1.5GHz. The differences of subcarrier-amplitudes make an effective of 0.3 bit reduction of soft decision bits of Viterbi decoder, and when the effective of 0.3 bit reduction can cause 0.5dB SNR degradation. This paper proposes two modem architectures which compensate for the degraded subcarrier by multiplying the reciprocal of degraded values in analog or digital domain. It is shown that, for the proposed architecture applied to MB-OFDM UWB, the performance improvements up to 0.5dB can be obtained over the conventional uncompensated receiver architecture.

**Key words** — MB-OFDM, UWB, Spectrum Equalizer

## I. INTRODUCTION

The UWB suffers heavy band degradation, because the path loss of a signal power is proportionate to the square of the signal's frequency as in (1).

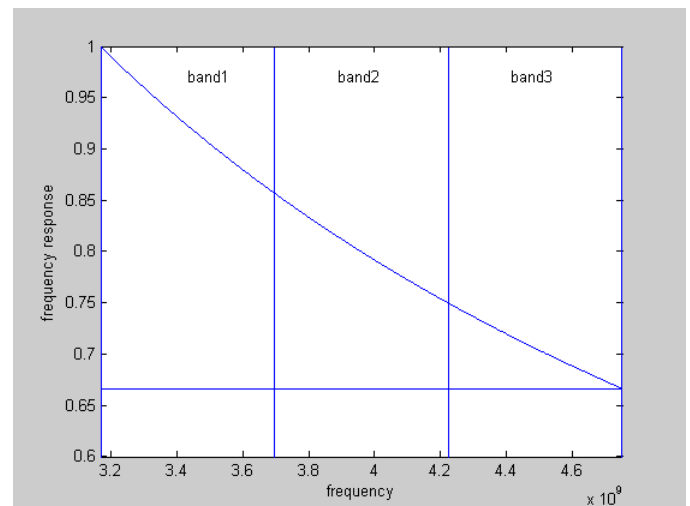
$$PL = \left(\frac{4\pi d}{\lambda}\right)^2 = \left(\frac{4\pi d}{c/f}\right)^2 = \left(\frac{4\pi df}{c}\right)^2 \quad (1)$$

$$= \left(\frac{4\pi d}{c}\right)^2 f^2$$

In (1),  $d$  is a distance of between transmit and receive antenna,  $\lambda$  is wavelength of used signal,  $c$  is the speed of light, and  $f$  is the frequency of the signal.

So, there can be existing slightly degraded and heavily diminished frequency parts. Because OFDM system carries information on frequency subcarrier, frequency degradation

means reduced input signal level of viterbi decoder. When a viterbi decoder received highly defected subcarrier as an input, the decoder can't use any information in MSB part of soft decision bits of subcarrier. This makes the viterbi decoder's gain reduced and the reduced decoder's gain increases bit error probability of the system.



**Fig. 1 Band degradation according to the path loss**

Fig. 1 shows band frequency response according to the path loss on mandatory channel of MB-OFDM UWB, 3.168~4.762GHz as in [2].

This paper proposes an analog and a digital compensators to reduce the effect of soft decision bit reduction.

## II. PROPOSED COMPENSATOR

### A. Band Degradation Effects

Received signals, suffered for band degradation as in fig. 1, have smaller average signal level than that of original one. In (2) and (3), the average subcarrier signal level normalized by the best one is described.  $X(w)$  is received signal level, and  $P(w)$  is received signal power level. And,  $f_{min}$ ,  $f_{max}$  are minimum and maximum frequency in band, respectively.

In case of MB-OFDM UWB mandatory channel, the average signal power and the voltage level are reduced to 67% and 81%, respectively. If N-bit soft decision is used, the average used bits in soft decision can be described as (4).

Before we show the simulation results, we need to consider synchronization of the receiver system. In this letter, we assume that the synchronization is perfect. The CSS signals consist of low- and high-band signals. If input signal is a high-band signal, high-band filter is designed to be operated, and if input signal is a low-band signal, low-band filter is also designed to be executed.

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Fig. 2 shows noise power and bandwidth of the filter in both of the conventional channel filtering method and the dual-band filtering method. The dual-band filtering method has a smaller noise power and bandwidth in the filter than the conventional channel filtering method, because the low- and high-band filter alternatively applied to the received signals at high- and low-bands.

When 3bit soft decision is used in a Viterbi decoder, the effective soft decision bits are reduced to 2.7 bits, so that the reduction of effective soft decision bits can make 0.5~1dB performance degradation. IEEE 802.15 low rate alternative PHY Task Group (TG 4a) for Wireless Personal Area Networks (WPANs) standardizes a new technique whose objective is in providing low rate communications, high precision ranging/location capability (1 meter accuracy and better), ultra low power, low cost, and etc. Therefore, a new proposal which consists of Ultra Wide Band -Impulse Radio (UWB-IR) operating in unlicensed UWB spectrum and Chirp Spread Spectrum (CSS) operating in unlicensed 2.4 GHz spectrum was adopted as a new standard technique by the committee of the IEEE 802.15 TG 4a in [1]. The CSS based on ISM band uses chirp signals categorized as spread spectrum signals and can afford to get a high precision ranging because of the good correlation properties of chirp signals. It uses bi-orthogonal codes for channel coding and has low hardware complexity by using differential encoding. Also, Simultaneously Operating Piconet (SOP) for multiple users is proposed by using different combination of sub-chirps in the CSS systems as in [1].

$$\begin{aligned} E\left[\frac{|P(w)|}{|P(f_{min})|}\right] &= \frac{f_{min}^2}{f_{max} - f_{min}} \int_{f_{min}}^{f_{max}} \frac{1}{f^2} df \\ &= \frac{f_{min}^2}{f_{max} - f_{min}} \left( \frac{1}{f_{min}} - \frac{1}{f_{max}} \right) \\ &= \frac{f_{min}}{f_{max}} \end{aligned} \quad (2)$$

$$\begin{aligned} E\left[\frac{|X(w)|}{|X(f_{min})|}\right] &= \frac{f_{min}}{f_{max} - f_{min}} \int_{f_{min}}^{f_{max}} \frac{1}{f} df \\ &= \frac{f_{min}}{f_{max} - f_{min}} (\ln f_{max} - \ln f_{min}) \\ &= \frac{f_{min}}{f_{max} - f_{min}} \ln \left( \frac{f_{max}}{f_{min}} \right) \end{aligned} \quad (3)$$

$$AveUsedSoftBits = \log_2(0.81 \cdot 2^N) = N - 0.30 \quad (4)$$

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### B. Analog Equalizer

Path loss is a function of the frequency and the distance as in (1). Because the degradation according to the distance is adapted to all bands of frequencies, signals can be controlled through automatic gain control (AGC). Additionally,  $f$  is the frequency of bands that we have already known.

If the received signal is equalized with a filter whose frequency response is proportionate to  $f$ , each frequency component of received signal can be even. Because three subbands hop frequency periodically and the hopped frequency signals are demodulated in receiver, RF equalizer is added to between LNA, which outputs all bands of frequencies, and Demodulator, which down converts RF signal into baseband signal. Because the input signals of equalizer contain AWGN as well as original signals, the equalizing of all bands can not get SNR improvement. But we can make the input signal levels of viterbi decoder even by compensation, and the even input level can minimize the bits loss of effective soft decision. Fig. 2 describes a structure of the MB-OFDM UWB modem containing an analog equalizer.

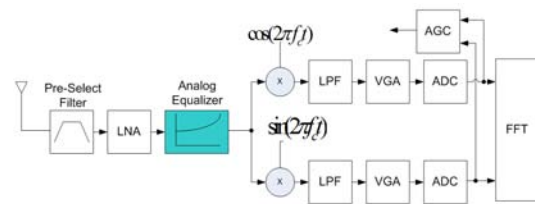


Fig. 1 UWB block diagram with Analog Equalizer

### C. Digital Compensator

The effects of band degradation can also be compensated in baseband. Because FFT module exists already in baseband of OFDM system, compensation can be made by multiplying path loss to frequency domain signals after FFT operation. There is not only AWGN but also quantization noise in digital compensator input, so that the amplified quantization noise makes system performance worse than that of analog one. But quantization noise can be ignored when SNR is moderately small. In case of (B+1) bit resolution analog-to-digital converter, the quantization noise of NSR is as in (5)[3].

$$NSR = -6.02B - 10.8 + 20 \log\left(\frac{X_m}{\sigma_x}\right) \quad (5)$$

In (5),  $X_m$  is the full-scale amplitude of ADC and  $\sigma_x$  is the root mean square(RMS) value of the signal amplitude. In case of 0 dBw signal power and 5bit ADC, noise power is less than -24dBw.

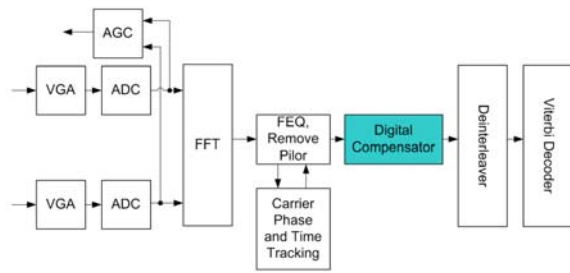


Fig. 2 UWB Modem Block Diagram with Digital Compensator

Fig. 2 shows modem architecture with digital compensator.

### III. SIMULATION RESULTS

Proposed architecture is applied to MB-OFDM UWB which has been proposed as a standard of IEEE 802.15.3a [4]. Simulation is performed in 480 Mbps mode using 3bit soft decision Viterbi decoder. Simulation environments and assumptions are listed in table I.

Table. 1

Environments	
System version	v.0891
ADC bits	5
ADC Dynamic range	7dB
Data rate	480Mbps
Channel decoder	Viterbi 3bit soft decision
Packet length	1024byte
Simulated packets	200
Assumption	
Path Loss	$\propto f^2$
Synch. & Channel Eq.	Perfect
multipath	X

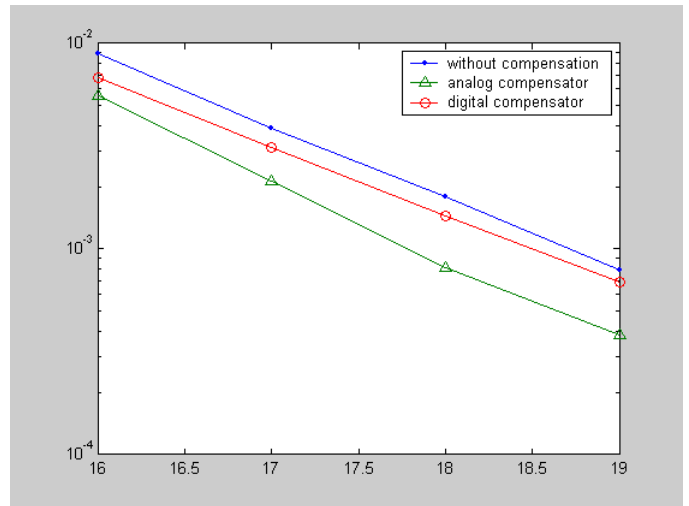


Fig. 3 BER performance of compensator

The BER and PER simulations of the proposed and the conventional channel filtering method was performed by using the Kiser-window filter in [2] which has 23 taps and 11 taps, respectively. This means that the hardware complexity of the proposed method was similar to that of the conventional method and we only compare the BER and PER performances of the proposed method with that of conventional method. Though an additional control logic that decides which one of the channel filters between high- and low-filter is used is necessary to the proposed method, it is very simple. On the other hand, power consumption of the proposed method is slightly better than that of the conventional method. Because the proposed method uses 11-tap filter for the received signal while the conventional method uses 23-tap filter in every time.

#### IV. CONCLUSION

This paper proposed a channel compensator which can maximize Viterbi decoder gain by the compensation of degraded subcarrier. As the simulation results, the proposed analog compensator can make 0.5~1dB SNR improvements at  $10^{-3}$  BER. Digital compensator can be used when the quantization noise is much smaller than AWGN. 5bit ADC is sufficient for 480Mbps mode of MB-OFDM. The SNR gain can be used to extend communication distance or to improve BER performance.

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