ABSTRACT

We report on the MIT Epoch of Reionization (MITEoR) experiment, a pathfinder low-frequency radio interferometer whose goal is to test technologies that improve the calibration precision and reduce the cost of the high-sensitivity 3D mapping required for 21 cm cosmology. MITEoR accomplishes this by using massive baseline redundancy, which enables both automated precision calibration and correlator cost reduction. We demonstrate and quantify the power and robustness of redundancy for scalability and precision. We find that the calibration parameters precisely describe the effect of the instrument upon our measurements, allowing us to form a model that is consistent with $\chi^2$ per degree of freedom $< 1.2$ for as much as 80% of the observations. We use these results to develop an optimal estimator of calibration parameters using Wiener filtering, and explore the question of how often and how finely in frequency visibilities must be reliably measured to solve for calibration coefficients. The success of MITEoR with its 64 dual-polarization elements bodes well for the more ambitious Hydrogen Epoch of Reionization Array (HERA) project and other next-generation instruments, which would incorporate many identical or similar technologies.

Key words: Cosmology: Early Universe – Radio Lines: General – Techniques: Interferometric – Methods: Data Analysis

1 INTRODUCTION

Mapping neutral hydrogen throughout our universe via its redshifted 21 cm line offers a unique opportunity to probe the cosmic “dark ages,” the formation of the first luminous objects, and the epoch of reionization (EoR). A suitably designed instrument with a tenth of a square kilometer of collecting area will allow tight constraints on the timing and duration of reionization and the astrophysical processes that drove it [Pober et al. 2014]. Moreover, because it can
map a much larger comoving volume of our universe, it has
the potential to overtake the Cosmic Microwave Background
(CMB) as our most sensitive cosmological probe of inflation,
dark matter, dark energy, and neutrino masses. For example,
Mao et al. [2008], a radio array with a square kilome-
ter of collecting area, maximal sky coverage, and good fore-
ground maps could improve the sensitivity to spatial curva-
ture and neutrino masses by up to two orders of magnitude,
∂Ωs ≈ 0.0002 and ∆mν ≈ 0.007 eV, and shed new light
on the early universe by a 4σ detection of the spectral index
running predicted by the simplest inflation models favored
by the BICEP2 experiment [Ade et al. 2014].

Unfortunately, the cosmological 21 cm signal is so faint
that none of the current experiments around the world (LO-
FAR [Röttgering 2003], MWA [Tingay et al. 2013], PAPER
Parsons et al. 2010, 21CMA [Wu 2009], GMRT [Paciga et al.
2011] have detected it yet, although increasingly stringent
upper limits have recently been placed [Paciga et al. 2013,
Dillon et al. 2014] Parsons et al. 2014]. A second challenge is
that foreground contamination from our galaxy and extra-
galactic sources is perhaps four orders of magnitude larger
than the cosmological hydrogen signal [de Oliveira-Costa
et al. 2008]. Any attempt to accurately clean it out from the
data requires even greater sensitivity as well as more accu-
rate calibration and beam modeling than the current state-
of-the-art in radio astronomy (see Furlanetto et al. 2006,
Morales & Wyithe 2010) for reviews).

Large sensitivity requires large collecting area. Since steerable single dish radio telescopes become prohibitively
expensive beyond a certain size, the aforementioned ex-
periments have all opted for interferometry, combining N
(generally a large number) independent antenna elements
which are (except for GMRT) individually more affordable.
The LOFAR, MWA, PAPER, 21CMA and GMRT exper-
iments currently have comparable N. The problem with scal-
ing interferometers to high N is that all of these experi-
ments use standard hardware cross-correlators whose cost
grows quadratically with N, since they need to correlate all
N(N − 1)/2 ∼ N^2/2 pairs of antenna elements. This cost
is reasonable for the current scale N ∼ 10^4, but will com-
pletely dominate the cost for N ∼ 10^5, making precision
cosmology arrays with N ∼ 10^6 as discussed in Mao et al.
(2008) insufficient in the near future, which has motivated
novel correlator approaches such as Morales [2011].

For the particular application of 21 cm cosmology, how-
ever, designs with better cost scaling are possible, as de-
scribed in [Tegmark & Zaldarriaga 2009, 2010]: by arranging
the antennas in a hierarchical rectangular or hexagonal grid
and performing the correlations using Fast Fourier Trans-
forms (FFTs), thereby cutting the cost scaling to N log N.
This is particularly attractive for science applications requir-
ing exquisite sensitivity at vastly different angular scales,
such as 21 cm cosmology (where short baselines are needed
to probe the cosmological signal) and long baselines are
needed for point source removal). Such hierarchical grids
thus combine the angular resolution advantage of traditional
array layouts with the cost advantage of a rectangular Fast
Fourier Transform Telescope. If the antennas have a broad
spectral response as well and their signals are digitized with
high bandwidth, the cosmological neutral hydrogen gets si-
multaneously imaged in a vast 3D volume covering both
much of the sky and also a vast range of distances (corre-
ducing to different redshifts, i.e., different observed fre-
quencies.) Such low-cost arrays have been called omniscopes
[Tegmark & Zaldarriaga 2009, 2010] for their wide field of
view and broad spectral range.

Of course, producing such scientifically rich maps with
any interferometer depends crucially on our ability to pre-
cisely calibrate the instrument, so that we can truly un-
derstand how our measurements relate to the sky. Tradi-
tional radio telescopes rely on a well-sampled Fourier plane
to perform self-calibration using the positions and fluxes of
a number of bright point sources. At first blush, one might
think that any highly-redundant array would be at a dis-
advantage in its attempt to calibrate the gains and phases
of individual antennas. However, we can use the fact that
redundant baselines should measure the same Fourier com-
ponent of the sky to improve the calibration of the array
dramatically and quantifiably. In fact, we find that the ease
and precision of redundant baseline calibration is a strong
rationale for building a highly-redundant array, in addition
to the improvements in sensitivity and correlator speed.

Redundant calibration is useful both for current gener-
ation redundant arrays like MITEoR and PAPER and for
future large arrays that will need redundancy to cut down
correlator cost. Omniscopes must be calibrated in real time,
because they do not compute and store the visibilities mea-
sured by each pair of antennas, but effectively gain their speed advantage by averaging redundant baselines in real
time. Individual antennas therefore cannot be calibrated in
post-processing. No calibration scheme used on existing low
frequency radio interferometers has been demonstrated to
meet the speed and precision requirements of omniscopes.
Thus, the main goal of the MIT Epoch of Reionization ex-
periment (MITEoR) and this paper is to demonstrate a suc-
cessful redundant calibration pipeline that can overcome the
calibration challenges faced by current and future generation
instruments by performing automatic precision calibration
in real time.

Building on past redundant baseline calibration meth-
ods by Wieringa [Wieringa 1992] and others, some of us re-
cently developed an algorithm which is both automatic and
statistically unbiased, able to produce precision phase and
gain calibration for all antennas in a hierarchical grid (up to
a handful of degeneracies) without making any assumptions
on the cosmological power spectrum have contributions from both
noise and sample variance, and it is well-known that the total
error bars on a given physical scale (for a fixed experimental cost)
are minimized when both contributions are comparable, which
happens when the S/N ∼ 1 on that scale. This is why more com-
pact 21 cm experiments have been advocated. This is also why
early suborbital CMB experiments focused on small patches of
sky to get S/N ∼ 1 per pixel, and why galaxy redshift surveys
target objects like luminous red galaxies that give S/N ∼ 1 per
3D voxel.
about the sky signal \cite{Liu2010}. Once obtained, precision calibration solutions can in turn produce more accurate modeling of the synthesized and primary beam \cite{Pober2012}, which has been shown to improve the quality of the foreground modeling and removal which is so crucial to 21 cm cosmology. It is therefore timely to develop a pathfinder instrument that tests how well the latest calibration ideas works in practice.

MITEoR is such a pathfinder instrument, designed to test redundant baseline calibration. We developed and successfully applied a real-time redundant calibration pipeline to data we took with our 64 dual-polarization antenna array during the summer of 2013 in The Forks, Maine. The goal of this paper is to describe the design of the MITEoR instrument, demonstrate the effectiveness of our redundant baseline calibration and absolute calibration pipelines, and use the calibration results to obtain an optimal scheme for estimating calibration parameters as a function of time and frequency.

This paper is organized as follows. We first describe in Section 2 the instrument, including the custom developed analog components, the 8 bit 128 antenna-polarization correlator, the deployment, and the observation history. In Section 3 we focus on precision calibration. We explain and quantitatively evaluate relative redundant calibration, and address the question of how often calibration coefficients should be updated. We also examine the absolute calibration, including breaking the degeneracies in relative calibration, mapping the primary beam, and measuring the array orientation. In Section 4, we summarize this work and discuss implications for future redundant arrays such as HERA \cite{Pober2014}.

2 THE MITEoR EXPERIMENT

In theory, a very large omniscope can be built following the generalized architecture in Figure 2. On the other hand, it is crucial to demonstrate that automatic and precise calibration is possible in real-time using redundant baselines, since the calibration coefficients for each antenna must be updated frequently to allow the FFTs to combine the signals from the different antennas without introducing errors. In this section, we will present our partial implementation of this general design, including both the analog and the digital systems. Because the digital hardware is powerful enough to allow it, the MITEoR prototype correlates all 128 input channels with one another, rather than just a small sample as mentioned in the caption of Figure 1. This provides additional cross-checks that greatly aid technological development, where instrumentation may be particularly prone to systematics. This also allows us to explore the question of exactly how often and how finely in frequency we must measure visibilities to solve for calibration coefficients, a question we return to in Section 3. Since we chose to implement a full correlator, an additional FFT correlator would bring no extra information (simply computing the same redundant-baseline-averaged visibilities faster), so we leave the digital implementation of an FFT correlator to future work.

In general, our mission is to empirically explore any challenges that are unique to a massively redundant interferometer array. Once these are known, one can reconfigure the cross-correlation hardware to perform spatial FFTs, thereby obtaining an omniscope with $N \log N$ correlator scaling.
2.1 The Analog System

MITeOR contains 64 dual-polarization antennas, giving 128 signal channels in total. The signal picked up by the antennas is first amplified by two orders of magnitude in power by the low noise amplifiers (LNAs) built-in to the antennas. It is then phase switched in the swapper system, which greatly reduces cross-talk downstream. The signal is then amplified again by about five orders of magnitude in the line-drivers before being sent over 50 meter RG6 cables to the receivers. The receivers perform IQ demodulation on a desired 50 MHz band selected between 100 MHz and 200 MHz, producing two channels with adjacent 25 MHz bands, and sends the resulting signals into the digitization boards containing 256 analog-to-digital converters (ADCs) sampling at 50 MHz. The swappers, line-drivers and receivers we designed are shown in Figure 2.

When designing the components of this system, we chose to use commercially-available integrated circuits and filters whenever possible, to allow us to focus on system design and construction. In some cases (such as with the amplifiers) the cost of the IC is less than the cost of enough discrete transistors to implement even a rough approximation of the same functionality. Less expensive filters could be made from discrete components, but the characteristics of purchased modules tend to be better due to custom inductors and shielding. When we needed to produce our own boards as described below, our approach was to design, populate and test them in our laboratory, then have them affordably mass-produced for us by Burns Industries.

2.1.1 Antennas

The dual-polarization antennas used in MITeOR were originally developed for the Murchison Widefield Array (Lonsdale et al. 2009; Tingay et al. 2013), and consist of two “bow-tie”-shaped arms as can be seen in Figure 3. They are inexpensive, easy to assemble, and sensitive to the entire band of our interest. The MWA antennas were designed for the frequency range 80-300 MHz, and have a built-in low noise amplifier with 20 dB of gain. The noise figure of the amplifier is 0.2 dB, and the 20 dB of gain means that subsequent gain stages do not contribute significantly to the noise figure.

2.1.2 Swappers (Phase Switches)

As with many other interferometers, crosstalk within the receivers, ADCs, and cabling significantly affects signal quality. We observe the cross-talk to depend strongly on the physical proximity of channel pairs, reaching as high as about −30 dB between nearest neighbor receiver channels. Our swapper system is designed to cancel out crosstalk during the correlator’s time averaging by selectively inverting analog signals using Walsh modulation (Sanchez 2011). The signal from each antenna-polarization is inverted 50% of the time according to its own Walsh function, by an analog ZMAS-1 phase switch from Mini-Circuits located before the second amplification stage (line-driver), then appropriately re-inverted after digitization. We perform the inversion once every millisecond, which is much longer than the ADC’s 20 ns sample time, and much shorter than the averaging time of a few seconds. This eliminates all crosstalk to first order (Sanchez 2011). If crosstalk reduction were the only concern, the ideal position for the swapper would be immediately after the antenna, in order to cancel as much crosstalk as possible. In practice, the swapper introduces a loss of about 3 dB, so we perform the modulation after the LNA to avoid adding noise (raising the system temperature). To evaluate the effectiveness of the swapper modules, we sent a monotone signal into one single channel of the receivers while leaving other channels open, and measured the correlation between the signal channel and each empty channels with the swapper turned on and off. We then repeated this while varying the signal frequency over the full range of interest. As seen in Figure 4 the swapper system attenuates crosstalk in the receiver and ADC by as much as 50 dB over the frequency band of interest, typically reducing it to being of order −80 dB for strongly afflicted signal pairs.

2.1.3 Line-driver

A line-driver (Figure 5) amplifies a single antenna’s signal from one of its two polarization channels while also powering its LNA. Line-drivers only handle a single channel to reduce potential crosstalk from sharing a printed circuit board. They are placed within a few meters of the antennas in order to reduce resistive losses from powering the antenna at low voltage. Additional gain that they provide early in the analog chain helps the signal overpower any noise picked up immediately after the antenna, in order to cancel as much crosstalk as possible.

Figure 2. System diagram of the analog system. The signal received with an MWA “bow-tie” antenna is first amplified by the built-in low noise amplifier, then Walsh-modulated in the swapper module controlled by the swapper system. The signal is amplified again in the line driver and sent to the processing rack through 50 m long coaxial cables. In the processing rack, the signal first goes into the receiver, where it undergoes further amplification, frequency down-mixing and I/Q modulation from the 120-180 MHz range to the 0-25 MHz range. The analog chain ends with digitization on ADC connected to ROACH boards.

3 http://www.burnsindustriesinc.com
4 In a multi-stage amplifier, the contribution of each stage’s noise figure is suppressed by a factor that is equal to the total gain of previous stages.
5 Since the undesirable crosstalk signal is demodulated with a different Walsh function than it is modulated with, it will be averaged out due to orthogonality of Walsh functions.
6 The inversion cannot be too frequent, because we need to discard data during the analog inversion process which takes a few microseconds. At the same time, the inversion needs to be frequent enough to average out the cross-talk.
Figure 4. Plots of cross-talk power measured in the laboratory. The swapper suppresses crosstalk between channels by as much as 50 dB. To measure these curves we fed a 0 dBm sinusoidal signal into input channel 0 of the receivers and left the other 31 input channels open. We then measured the correlations between channel 0 and all 31 empty channels, due to crosstalk from channel 0. We repeated the procedure with input frequencies from 125-150 MHz and obtained the results shown above.

Figure 3. System diagram of our swapper signal system and physical components of the swapper transceiver (lower left) and swapper controller (lower right). The swapper is designed to reduce crosstalk between neighboring channels.

Figure 5. System diagram and physical components of the line drivers. The line driver we designed takes the signal in the 50Ω coaxial cable from the antenna LNA and amplifies it by 51 dB, in order to overpower noise picked up in the subsequent 75Ω coaxial cable and further processing steps up to 50 meters away. It operates on 5V DC and also provides DC bias power to the antenna’s LNA through the 50 Ohm cable.

2.1.4 Receiver

Our receivers (Figure 6) take input from the line-drivers, bandpass filter the incoming signals, amplify their power level by 23 dB, and IQ-demodulate them. The resulting signals go directly to an ADC for digitization. Receivers are placed near the ADCs to which they are connected to reduce cabling for local oscillator (LO) distribution and ADC connections. IQ demodulation is used, which doubles received bandwidth for a given ADC frequency at the cost of using two ADC channels, and has the advantage of requiring only a single LO and low speed ADCs. The result is 40 MHz of us-
able bandwidth anywhere in the range 110-190 MHz, with a 2-3 MHz gap centered around the LO frequency due to bandpass filters. The receiver boards have five pins allowing their signals to be attenuated by any amount between 0 dB and 31 dB (in steps of 1 dB) before the second amplification stage, to avoid saturation and non-linearities from RFI and to attain signal levels optimal for digitization.

2.2 The Digital System

We designed MITEoR’s digital system (Figure 7) to be highly compact and portable. The entire system occupies 2 shock-mounted equipment racks on wheels, each measuring about 1 m on all sides. It takes in data from 256 ADC channels (64 antennas with I and Q signals for polarizations), Fourier transforms each channel, reconstructs IQ demodulated channels back to 128 corresponding antenna channels, computes the cross-correlations of all pairs of the 128 antenna channels with 8 bit precision, and then time-averages these cross-correlations. Although standard 4 bit correlators suffice for most astronomical observation tasks, the better dynamic range of our 8 bit correlator allows us to observe faint astronomical signals at the same time as 10^3 times brighter ORBCOMM satellites, whose enormous signal-to-noise has proved invaluable in characterizing various aspects of the system (see Sections 3.2.2, 3.2.3, and 3.3). The digital hardware is capable of processing an instantaneous bandwidth of 12.5 MHz with 49 kHz frequency bins. It averages those correlations and then writes them to disk every few seconds (usually either 2.6 or 5.3 seconds).

While one of the advantages of a massively redundant interferometer array is the ability to reduce costs by performing a spatial FFT rather than a full cross-correlation, we have not implemented FFT correlation in the current MITEoR prototype as the hardware is powerful enough to correlate all antenna pairs in real time (the feasibility of implementing FFT correlation on the ROACH platform has been demonstrated by Foster et al. (2014)). Rather, the goal of MITEoR is to quantify the accuracy that automatic redundant baseline calibration can attain, thereby experimentally characterizing all of the unknowns in the system, such as unexpected analog chain systematics and other barriers to finding good calibration solutions.

We adopted the widely-used F-X scheme in MITEoR’s digital system. We have 4 synchronized F-engines that take in data from 4 synchronized 64-channel ADC boards, which run at 12 bits and 50Ms/s. The F-engines perform the FFT and IQ reconstruction, and distribute the data onto 4 X-engines through 16 10GbE links. The 4 asynchronous X-engines each perform full correlation on 4 different frequency bands on all 128 antenna polarizations, and send the time averaged results to a computer for data storage.

To implement the computational steps of the MITEoR design, we used Field Programmable Gate Arrays (FPGAs). These devices can be programmed to function as dedicated pieces of computational hardware. Each F-engine and X-engine is implemented by one Xilinx FPGA (Virtex-5 for F-engines and Virtex-6 for X-engines). These FPGAs are seated on custom hardware boards developed by the CASPER collaboration (Parsons et al. 2006). We also use the software tool flow developed by CASPER to design...
the digital system. The CASPER collaboration is dedicated to building open-source programmable hardware specifically for applications in astronomy. We currently use two of their newer devices, the ROACH$^9$(Reconfigurable Open Architecture Computing Hardware) for the F-engines, and the ROACH$^9$ for the X-engines. The main benefit of using CASPER hardware is that it facilitates the time-consuming process of designing and building custom radio interferometry hardware. The CASPER collaboration also offers a large open-source library of FPGA blocks for commonly used signal processing structures such as polyphase filter banks, FIR filters and fast Fourier transform blocks (Parsons 2009). However, due to MITEoR’s ambitious architecture, involving both extreme compactness, an 8-bit correlator, and tight inter-ROACH synchronization constraints, we custom-designed most of the digital FPGA blocks. The specifications of our latest correlator are listed in Table 1.

| Antenna          | MWA dual-pol bow-tie |
|------------------|----------------------|
| Antenna count    | 64 × 2 polarizations |
| Array configuration | 8 × 8 grid          |
| ADC              | 4 × 64-channel 50 Msps |
| F-engine         | 4 ROACHes with Virtex-5 |
| X-engine         | 4 ROACH2s with Virtex-6 |
| Correlator precision | 8 bits               |
| Frequency range  | 110-190 MHz          |
| Instantaneous bandwidth | 12.5 MHz (50 MHz digitized) |
| Frequency resolution | 49 kHz               |
| Time resolution  | ≥ 2.68 s             |

Table 1. List of MITEoR specifications. We observed with two different 8 by 8 array configurations, one with 3 m separation and one with 1.5 m separation. We observed ORBCOMM band with 2.68 s resolution, and we chose a resolution of 5.37 s for other bands.

The MITEoR experiment was designed to be portable and easy to assemble. The entire experiment was loaded into a 17 foot U-Haul truck and driven to The Forks. It took a crew of 15 people less than 2 days to assemble the instrument and bring it to full capacity. A skeleton crew of 3 members stayed on site for monitoring and maintenance for the following two weeks, during which we collected more than 300 hours of data. Subsequently, a demolition crew of 5 members disassembled and packed up MITEoR in 6 hours and concluded the successful deployment.

During the deployment, we scanned through the frequency range 123.5-179.5 MHz, with at least 24 consecutive hours at each frequency. We used two different array layouts for most of the frequencies we covered. The observation began with the antennas arranged in a regular 8 by 8 grid, with 3 meter spacing between neighboring antennas, which we later reconfigured to an 8 by 8 regular grid with 1.5 meter spacing for a more compact layout (which provides better signal-to-noise ratio on the 21 cm signal). The total volume of binary data collected was 3.9TB, and in the rest of this paper, we demonstrate the results of our various calibration techniques using this data set.

### 2.3 MITEoR Deployment and Data Collection

We deployed MITEoR in The Forks, Maine, which our online research suggested might be one of the most radio quiet regions in the United States at the frequencies of interest (Bowman & Rogers 2010a). We deployed the first prototype in September 2010, and performed a successful suite of test observations with an 8-antenna interferometer. In May 2012, we completed and deployed a major upgrade of the digital system to fully correlate $N = 16$ dual-polarization antennas. With the experience of this successful deployment, we further upgraded the digital system to accommodate $N = 64$ dual-polarization antennas, which led to our latest deployment in July 2013 and the results we describe in this paper.

9 https://casper.berkeley.edu/wiki/ROACH
10 https://casper.berkeley.edu/wiki/ROACH-2_Revision_2
11 The Forks has also been successfully used to test the EDGES experiment (Bowman & Rogers 2010b), and we found the RFI spectrum to be significantly cleaner than at the National Radio Astronomy Observatory in Green Bank, West Virginia at the very low (100-200 MHz) frequency range that is our focus: the entire spectrum at The Forks is below -100 dBm except for one -89.5 dBm spike at 150 MHz.

### 3 CALIBRATION RESULTS

As we have emphasized above, the precision calibration of an interferometer is essential to its ability to detect the faint cosmological imprint upon the 21 cm signal, and the key focus of MITEoR is to determine how well real-time redundant calibration can be made to work in practice. In this section we describe the calibration scheme that we have designed and implemented and quantify its performance. We first constrain the relative calibration between antennas, utilizing both per-baseline algorithms and redundant-baseline calibration algorithms (Liu et al. 2010). We then build on these relative calibration results to constrain the absolute calibration of the instrument, including breaking the few degeneracies inherent to redundant calibration.

12 We aligned the antenna positions using a laser-ranging total station, and measured their positions with millimeter level precision. The median deviation from a perfect grid is 2 mm in the N-S direction, 3 mm E-W, and 28 mm vertical, primarily caused by the fact that the deployment site had not been leveled.
3.1 Relative Calibration

3.1.1 Overview

The goal of relative calibration is to calibrate out differences among antenna elements caused by non-identical analog components, such as variations in amplifier gains and cable lengths, which may be functions of time and frequency. We parametrize the calibration solution as a time- and frequency-dependent multiplicative complex gain \( g_i \) for each of the 128 antenna-polarizations. Calibrating the interferometer amounts to solving for the coefficients \( g_i \) and undoing their effects on the data. Our calibration scheme revolves around calibration methods that heavily utilize the redundancy of our array, whose efficacy we aim to demonstrate with MITEoR. The current redundant calibration pipeline involves three steps, as illustrated in Figure 9:

(i) Rough calibration computes approximate calibration phases using knowledge of the sky.

(ii) Logarithmic calibration (“logcal”) decomposes roughly calibrated data into amplitudes and phases and computes least square fits for amplitude and phase separately.

(iii) Linear calibration (“lincal”) takes the relatively precise but biased results from logcal and computes unbiased calibration parameters with even higher precision.

Although logcal and lincal have been previously proposed (Wieringa 1992; Liu et al. 2010; Marthi & Chengalur 2014), they both fail in their original form if the phases of \( g_i \) are not close to 0.\(^{13}\) In practice, the phases of \( g_i \) can be anywhere in the interval \([0, 2\pi)\). To overcome these practical challenges, we introduced various improvements to these algorithms. In the following sections, we describe our improvements to calibration algorithms in detail, and demonstrate the effectiveness of our calibration by obtaining \( \chi^2/\text{DoF} \approx 1 \) for the majority of our data. We then analyze these calibration parameters to construct a Wiener filter to optimally average them over time and frequency, which also tells us how frequently we need to calibrate in time and frequency.

3.1.2 Rough Calibration

The goal of rough calibration is to obtain reliable initial phase estimates for the calibration parameters to enable the subsequent more sophisticated algorithms. This step does not have to involve redundancy, thus it can be done with any standard calibration techniques, for example self-calibration (Rau et al. 2009; Braun 2013). The rough calibration algorithm that we describe below is computationally cheap and can robustly improve upon raw data even when a few antennas have failed.

At a given time and frequency, we have both the measured visibilities, \( v_{ij} \), and \( v_{ij}^{\text{model}} \), a rough model of the true sky signal\(^{14}\) where indices \( i \) and \( j \) represent antenna number. We first compute the phase difference between each

\(^{13}\) Logcal requires phase calibrations close to 0 to avoid phase wrapping issues, whereas lincal requires phase calibrations close to 0 to converge.

\(^{14}\) Since we are trying to obtain an initial estimate, the model does not have to be very accurate.
measured visibility and its prediction. We then pick one reference antenna and subtract the phases of its visibilities from the phases of other visibilities to obtain a list of estimated phase calibration for each antenna. Finally, we take the median of these calibration phases to obtain a robust phase calibration estimator for each antenna. More concretely, we use the following procedure:

(i) Construct a matrix $M$ of phase differences where $M_{ij} = -M_{ji} = \arg(v_{ij}/v_{ij}^{\text{model}})$.

(ii) Define the first antenna as the reference by subtracting the first column of $M$ from all columns to obtain $M'_{ij} = M_{jk} - M_{j0}$.

(iii) Obtain rough phase calibration parameters $\phi_k \equiv \arg(g_k)$ by computing the median angle of column $k$ in $M'$, defined as

$$\phi_k \equiv \arg[\text{median}_i\{\exp(iM'_{ik})\}] = \arg[\text{median}_i\{\cos(M'_{ik})\} + i\text{median}_i\{\sin(M'_{ik})\}].$$

For stable instruments, the true calibration parameters have very small variation over days, so we can use one set of rough calibration parameters from a single snapshot in time for data from all other times. Thus we pick a snapshot at noon when each $v_{ij}^{\text{model}}$ can be easily computed from position of the Sun, and use the resulting raw calibration parameters as the starting point for logical at all other times.

### 3.1.3 Log Calibration and Linear Calibration

To explain our redundant calibration procedure, we first need to briefly reintroduce the formalism developed in Liu et al. (2010). Suppose the $i$th antenna measures a signal $s_i$ at a given instant. This signal can be expressed in terms of a complex gain factor $g_i$, the antenna’s instrumental noise contribution $n_i$, and the true sky signal $x_i$ that would be measured in the limit of perfect gain and no noise:

$$s_i = g_i x_i + n_i.$$  

Under the standard assumption that the noise is uncorrelated with the signal, each baseline’s measured visibility is the correlation between the two signals from the two antennas:

$$v_{ij} \equiv \langle s_i s_j \rangle = g_i^* g_j \langle x_i^* x_j \rangle + g_i^* \langle x_i^* n_j \rangle + g_j^* \langle n_i^* x_j \rangle + \langle n_i^* n_j \rangle$$

$$= g_i^* g_j y_{i-j} + n_{ij}^{\text{REF}},$$  

where we have denoted the true correlation $\langle x_i^* x_j \rangle$ by $y_{i-j}$,15 the noise from each antenna by $n_i$, the noise for each baseline by $n_{ij}^{\text{REF}}$, and expectation values (effectively time averages) by angled brackets $\langle \cdots \rangle$. In a maximally redundant array such as MITEoR, the number of unique baselines is much smaller than the total number of baselines. Therefore, we can treat all the $g_i$s and the $y_{i-j}$s as unknowns while keeping the system of equations overdetermined, enabling fits for both despite the presence of instrumental noise.

In Liu et al. (2010), some of us proposed logical and linear, and we have implemented both for calibrating MITEoR data. In log calibration, we take the logarithm of both sides of Equation 3 and obtain a linearized equation in logarithmic space. We then perform a least squares fit for the system of equations

$$\log v_{ij} = \log g_i^* + \log g_j + \log y_{i-j},$$  

where we solve for $\log g_i$ and $\log y_{i-j}$. Because the least squares fit takes place in log space whereas the noise is additive in linear space, the best fit results are biased. Linear calibration, on the other hand, is unbiased (Liu et al. 2010). The lincal method performs a Taylor expansion of Equation 3 around initial estimates $g_i^*$ and $y_{i-j}^*$ and obtains a system of linearized equations

$$v_{ij} = g_i^* y_j^* y_{i-j} + g_i^* y_j^* y_{i-j}^* + g_i^* y_j^* y_{i-j}^* + g_i^* y_j^* y_{i-j}^*,$$  

where we solve for $g_i^*$ and $y_{i-j}^*$. For a detailed description of the logical and lincal algorithms and their noise properties, we direct the reader to Liu et al. (2010); Marthi & Chengalur (2014). We now describe some essential improvements to these algorithms.

Logical was first thought to be unable to calibrate the phase component due to phase wrapping, since logcal has no way to recognize that 0° and 360° are the same quantity. Consider, for example a pair of redundant baselines that measure phases of 0.1° and 359.9° respectively. We can infer that they each only need a very small phase correction (±0.1°) to agree perfectly. However, since logical treats the difference between them as 359.8° rather than 0.2°, it will calibrate by averaging 0.1° and 359.9° to 180°, which is completely wrong.

We made two improvements to the logical method to guard against this. The first is to perform rough calibration beforehand, as described in Section 3.1.2. The second is to re-wrap the phases of $v_{ij}$. While rough calibration can make the phase errors relatively small,16 that improvement alone is not sufficient, since 0° and 360° are still treated as different quantities. Thus we need to intelligently wrap the phases of the input vector before feeding it into logical. This is done in two simple steps. For a snapshot of rough calibrated visibilities at given time and frequency, $v_{ij}$, we first estimate the true phases of each group of redundant baselines, $\arg(y_{i-j})$, by computing median angles of measured phases using Eq. 1. Then for each measured phase, we add or subtract $2\pi$ until it is within $\pm \pi$ of $\arg(y_{i-j})$. This eliminates the phase wrapping problem.

Unlike logical, lincal is an unbiased algorithm, but it relies on a set of initial estimates for the correct calibration solutions to start with. The output of lincal can be fed back into the algorithm and it can iteratively improve upon its own solution. However, the algorithm converges to the right answer only if the initial estimates are good. In practice, we find that three iterations of lincal typically produces excellent convergence, because the outputs of logical are already decent estimates of the calibration solutions. Thus, by improving logical, we also greatly improve lincal’s effectiveness.
Our current calibration pipeline performs all steps of redundant calibration in less than 1 millisecond on a single processor core for data slice at one time and one frequency channel, which is an order of magnitude faster than the rate data is saved onto disk. It is carried out by our open source Omniscope package, coded in C++/Python. Thus there should be no computational challenge in performing the above described calibration procedure in real-time for any array with less than \(10^3\) elements. For a future omniscope that has as many as \(10^6\) elements, there are two ways to reduce the computational cost. The first is to calibrate less frequently in time and frequency, and we will discuss in detail the minimal sampling frequency in Section 3.1.5. The other is to adapt a hierarchical redundant calibration scheme, where instead of calibrating all visibilities at the same time, one can calibrate the array in a hierarchical fashion whose computational cost scales only linearly with the number of elements. We discuss more details regarding hierarchical redundant calibration in Appendix B.

### 3.1.4 \(\chi^2\) and Quality of Calibration

One of the many advantages of redundant calibration is it allows for the calculation of a \(\chi^2\) for every snapshot to quantify how accurate the estimated visibilities are for each unique baseline, even without any knowledge of the sky. For a set of visibilities at a given time and frequency, \(v_{ij}\), with calibration results \(g_i\) and \(g_j\), we define \(\chi^2\) as

\[
\chi^2 = \sum_{ij} \frac{|v_{ij} - y_{ij}^* g_i^* g_j|^2}{\sigma_{ij}^2},
\]

where \(\sigma_{ij}^2\) is the noise contribution to the variance of the visibility \(v_{ij}\). The effective number of degrees of freedom (DoF) is

\[
\text{DoF} = N_{\text{measurements}} - N_{\text{parameters}} = N_{\text{baselines}} - (N_{\text{antennas}} + N_{\text{unique baselines}}).
\]

The numerator in Equation 6 represents the deviation of measured data, \(v_{ij}\), from the best fit redundant model, \(y_{ij}^* g_i^* g_j\). Thus \(\chi^2/\text{DoF}\) can be interpreted as the non-redundancy in measured data divided by the expected non-redundancy from pure noise. If the data agrees perfectly with the redundant model (with noise) and is free from systematics, then \(\chi^2/\text{DoF}\) is drawn from a \(\chi^2\) distribution with mean 1 and variance \(2/\text{DoF}\). With a smooth model for \(\sigma_{ij}\) which we describe below, we compute \(\chi^2/\text{DoF}\) for the results of rough calibration, log calibration, and linear calibration using all of our data. The \(\chi^2\) distributions of our calibrations for one day’s data are shown in Figures 10 and 11. Each calibration algorithm significantly reduces the \(\chi^2/\text{DoF}\), and lincal’s produces a distribution of \(\chi^2/\text{DoF}\) consistently centered around 1. We automatically flag any data with \(\chi^2/\text{DoF}\) larger than 1.2, which accounts for about 20% of the data. Among the data that is not flagged, 85% is accounted for by the theoretical \(\chi^2\) distribution. The 15% in the right tail is mostly attributable to a slightly optimistic noise model designed to avoid underestimating \(\chi^2\). This close agreement between predicted and observed \(\chi^2\) distributions for the lincal results suggests that except during periods that get automatically flagged, our instrument and analysis pipeline is free from significant systematic errors. The fully automatic nature of our calibration pipeline and data quality assessment is encouraging for future instruments with data volume too large for direct human intervention.

Calculating \(\chi^2/\text{DoF}\) for flagging and data quality assessment requires an accurate model of noise in the measured visibilities. To compute the noise \(\sigma_{ij}\), we approximate \(\sigma_{ij}^2\) by \(<\sigma^2>\), where the average is over all baselines. This assumption that all antennas have the same noise properties drastically decreases the computational cost of calculating...
χ²/DoF. Because we have 10⁷ baselines, and the variation of σij between baselines is less than 20% (due to slightly different amplifier gains), this approximation should cause only about a 1% error in the final χ²/DoF.

To compute ⟨σ²⟩, we perform linear regression on each visibility vij over one minute to obtain its estimated variance σij², and then average all σij² to obtain σ². Then we have ⟨σ²⟩ at all frequencies every minute. Before we plug ⟨σ²⟩ into Equation 8, we model it as a smooth and separable function: ⟨σ²⟩(f, t) = F(f)T(t), where F(f) and T(t) are polynomials. The smooth model has three advantages. The first is that it is physically motivated to model thermal fluctuation as a smooth and separable function. Secondly, a smooth noise model makes the χ²/DoF a much more sensitive flagging device. Theoretically, χ²/DoF should not rise above 1 when unwanted radio events such as radio frequency interference (RFI) occur, because they are far field signals that do not violate any redundancy. However, since RFI events make both the signal and noise stronger, by demanding a smooth noise model, the ⟨σ²⟩ we use will underestimate the noise during RFI events and give abnormally high χ²/DoF, which can then be successfully flagged with the χ²/DoF < 1.2 threshold. Thirdly, seasonal changes aside, the noise model is expected to largely repeat itself from day to day, so for future experiments that will operate for years, it suffices to use the same model repeatedly without recomputing σij in situ for all the data. Thus, by using a smooth noise model, one can drastically reduce the occurrence of false negatives (since it is better to flag good data than it is to fail to flag bad data) as well as the computational cost of calculating χ²/DoF.

3.1.5 Optimal Filtering of Calibration Parameters

While the above-mentioned estimates of the calibration parameters that we obtain from redundant baseline calibration vary over time and frequency, much of that variation is due to the noise in raw data. To minimize the effect of instrumental noise on the calibration parameters, we would like to optimally average information from nearby times and frequencies to estimate the calibration parameters for any particular measurement.

As we will show below, the optimal method for performing this averaging is Wiener filtering. In the rest of this section, we first measure the power spectrum of the calibration parameters over time and frequency, and make a determination of how to decompose this into contributions from signal (true calibration changes) and noise. We then weight the Fourier components in a way that is informed by their signal-to-noise ratio, and quantify how this Wiener filtering procedure improves upon more naive averaging over time and/or frequency. Finally, we discuss the implications for how regularly (in time and frequency) we should calibrate. It is important to note that while these methods are applied only to MITEoR below, they are applicable to any current or future experiment.

We model the measured calibration parameter gi(f, t) for the i⁰ antenna as the sum of a true calibration parameter si(f, t) (the “signal”) and uncorrelated noise ni(f, t):

\[ g_i(f, t) = s_i(f, t) + n_i(f, t). \]

We choose our estimator ˆgi(f, t) of the true calibration parameter si(f, t) to be a linear combination of the observed calibration parameters gi at different times and frequencies:

\[ ˆg_i(f, t) = \int \int W(f, f', t, t') g_i(f', t') df' dt' \]

for some weight function W. We optimize the estimator ˆgi by choosing the weight function W that minimizes the mean-squared estimation error ⟨∥g_i(f, t) − ˆg_i(f, t)∥²⟩. Assuming that the statistical properties of the signal and noise fluctuations are stationary over time, all correlation functions become diagonal in Fourier space:

\[ \langle \hat{s}_i(\tau, \nu) \hat{s}_i^*(\tau', \nu') \rangle = \langle 2\pi \rangle^2 \delta(\tau - \tau') \delta(\nu - \nu) S(\tau, \nu), \]

\[ \langle \hat{n}_i(\tau, \nu) \hat{n}_i^*(\tau', \nu') \rangle = \langle 2\pi \rangle^2 \delta(\tau - \tau') \delta(\nu - \nu) N(\tau, \nu), \]

\[ \langle \hat{s}_i(\tau, \nu) \hat{n}_i^*(\tau', \nu') \rangle = 0, \]

where tildes denote Fourier transforms and S and N are the power spectra of signal and noise, respectively. This means that the optimal filter becomes a simple multiplication ˆg = W ˆg in Fourier space, corresponding to the weighting function ˆW(τ, ν) that minimizes the mean-squared error

\[ \langle \| \hat{W}(\tau, \nu) \hat{g}_i(\tau, \nu) - \hat{s}_i(\tau, \nu) \|^2 \rangle. \]

Requiring the derivative of this with respect to ˆW to vanish gives the Wiener filter \( \hat{W}(\tau, \nu) = \frac{S(\tau, \nu)}{S(\tau, \nu) + N(\tau, \nu)} \).

We perform this analysis on data over 12 MHz and two hours, where the signal and noise power are empirically found to be approximately time-independent.
Since $S$ and $N$ are to a reasonable approximation independent of the antenna number $i$, we have dropped all subscripts $i$ for simplicity. Back in real space, the optimal estimator $\hat{g}_i$ for the $i$th calibration parameter is thus $g_i$ convolved with the 2D inverse Fourier transform of $\tilde{W}$.

To demonstrate this technique, we show the above process carried out in the time dimension in Figure 12. In practice we perform the analysis on time and frequency dimensions simultaneously through a 2D FFT. The noise power spectrum $N(\nu)$ is seen to be well fit by a combination of two power laws: $S(\nu) \approx (\nu/2.9 \times 10^{-5} \text{Hz})^{-2.7} + (\nu/4.8 \times 10^{-17} \text{Hz})^{-0.46}$. The optimal convolution kernel $W$ is seen to perform a weighted average of the data on the timescale of roughly 200 s and frequency scale of 0.15 MHz, giving the greatest weight to nearby times and frequencies, resulting in an order-of-magnitude noise reduction.

To quantify the effectiveness of the obtained filter compared to naive “boxcar” averages, we use the 2D power spectrum and noise floor of the calibration parameters obtained from real data to simulate many realizations of calibration parameters $g(f,t) = s(f,t) + n(f,t)$, apply various averaging/convolution schemes $W(f,t)$ on the simulated data, and compare their effectiveness by computing the RMS error $\langle |(W \ast g)(f,t) - s(f,t)|^2 \rangle$ normalized by $\langle |n|^2 \rangle$. Due to our limited frequency bandwidth as well as frequent RFI contamination, power spectrum modeling in the frequency dimension is very challenging, so the frequency Wiener filter appears to be less effective than the time filter. In Table 2 we list the normalized noise powers using frequency Wiener filter, time Wiener filter, 2D Wiener filter, as well as traditional boxcar averaging, and the 2D Wiener filter produces results three times less noisy than that of the traditional boxcar averaging.

We have described how to optimally average calibration parameters when we calibrate very regularly in time and frequency. For a future instrument such as an omniscope with $10^6$ antennas, calibration will pose a serious computational challenge, so it is important to know what the minimal frequency one needs to calibrate the instrument. The above analysis conveniently provides an answer to this question. As shown in the second panel of Figure 12 the signal is band limited. By the Nyquist theorem, one needs to sample with at least double the frequency of signal bandwidth, so in our case we could measure the calibration parameters without aliasing problems as long as we calibrate once per minute. Calibrating more frequently than this simply helps average down the noise. Although this one-minute timescale depends on the temporal stability characteristics of the amplifiers and other components used in our particular experiment, it provides a useful lower bound on what to expect from future experiments whose analog chains are even more stable.

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19 We only show results for amplitude calibration parameters for brevity, as the phase calibration results have nearly identical power spectrum.

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Figure 12. Illustration of 1D Wiener filtering of calibration parameters at different times. Panel (a) shows the amplitude of calibration parameters measured for one antenna over two hours. Panel (b) shows that the average power spectrum across all antennas (blue dots) is well fit by a white noise floor (red horizontal line) plus a sum of two power laws (green curve). Panel (c) shows the Wiener filter in frequency domain computed using Eq. 3.1.5 and the power spectra from panel (b). Panel (d) shows the Wiener convolution kernel in the time domain, the Fourier transform of the filter in Panel (c). Panel (e) shows the best estimates of the true calibration amplitude. The effectiveness of this filter is compared with that of other filters in Table 2.
Table 2. Wiener filtering reduces the noise contribution to the calibration parameters by an order of magnitude. This table lists residual noise power (normalized by original noise power) after applying various filters to average the amplitude calibration parameters in time and/or frequency. The optimal two-dimensional Wiener filter indeed performed the best, lowering the noise power by an order of magnitude. In comparison, the naive boxcar average, using the characteristic scales of the optimal Wiener filter (200 s and 0.15 MHz), has more than 3 times residual noise power than the Wiener filtered result.

| Averaging method                        | Relative noise power |
|-----------------------------------------|----------------------|
| No average                              | 1                    |
| Frequency Wiener filter                 | 0.33                 |
| Time Wiener filter                      | 0.12                 |
| Time and frequency Wiener filter        | 0.09                 |
| Time and frequency boxcar average       | 0.32                 |

3.2 Absolute Calibration

The absolute calibration of the array involves two separate tasks. One is to find the overall gain and to break phase degeneracies that redundant baseline calibration is unable to resolve, and the other to calibrate fixed properties of the instrument such as the orientation of the array and shape of the primary beam. The former is done by comparing the data to a sky model comprised of the global sky model (GSM) (de Oliveira-Costa et al. 2008) and published astronomical catalogs (see Jacobs et al. 2013 for example). The latter is done using bright point sources with known positions. While we can take advantage of the extremely high signal-to-noise data in the ORBCOMM channels (around 137 MHz), thanks to the dynamic range provided by our 8 bit correlator, it is important to note that all the algorithms described here are applicable to astronomical point sources as well.

This section is divided into three parts. The first part describes how we use prior knowledge of the sky to break the degeneracies in redundant calibration results, a vital step to obtain usable data products. The second and third part each describe one aspect of absolute calibration using satellite data: primary beam measurement and array orientation.

3.2.1 Breaking Degeneracies in Redundant Calibration

Redundant calibration alone cannot produce directly usable data products, due to the degeneracies intrinsic to the algorithms. There is one degeneracy in the amplitude of the calibration coefficients $g_i$, since scaling the amplitude of everything up by a common factor does not violate any redundancy (the degeneracies discussed here are per frequency and per time, as are the calibration solutions). There are three degeneracies in phase, corresponding to three degrees of freedom in a two dimensional linear field (see Appendix A for a detailed discussion). In general, breaking these degeneracies requires prior knowledge of the sky. In this section, we briefly describe our algorithm that uses the global sky model (GSM) of de Oliveira-Costa et al. (2008) to remove these degeneracies. Doing so requires efficiently simulating the response of the instrument to the GSM; we summarize a fast algorithm for doing so in Appendix C. We defer detailed comparison of our data and the GSM to a future publication.

Our degeneracy removal procedure is an iterative loop that repeats two steps. The first step is to fit for the amplitude degeneracy factor. The knowledge of the GSM and bright point sources give us a set of model visibilities, $\tilde{m}_{ij}$, where index $a$ denotes different modeled components such as the GSM or Cygnus A. A linear combination of these models should be able to fit our measurements. Thus we fit for the weights $w_a$ of the models by minimizing

$$|v_{ij} - \sum_a w_a \tilde{m}_{ij}|^2 .$$  \hspace{1cm} (13)

The second step is to break the degeneracy in redundant phase calibration by fitting for the degeneracy vector $\Phi$ and the constant $\psi$ defined in Appendix A. We assume that the error in the first step’s fitting is mostly due to the phase degeneracies, so we take the $w_a$ from step one and fit for $\Phi$ and $\psi$ by minimizing

$$|\arg(v_{ij}) - \arg(\sum_a w_a \tilde{m}_{ij}) - d_{i-j} \cdot \Phi - \psi|^2 ,$$ \hspace{1cm} (14)

where $d_{i-j}$ is the position vector for baseline $i - j$.

Note that the two fitting processes described above are not independent of one another, so we repeat these steps until convergence is reached. We find that in practice, the errors converge within two iterations. Our preliminary result is illustrated in Figure 13, which shows that the data agree very well with current models.

3.2.2 Beam Measurement Using ORBCOMM Satellites

In general, in situ measurements of antenna primary beams over large fields of view pose a challenge to 21 cm cosmology, as primary beam uncertainties are intimately related to calibration, imaging, and catalog flux uncertainties (Jacobs et al. 2013). Motivated by these difficulties, Pober et al. (2012) present a solution that uses celestial point sources and assumes reflection symmetry of the beam, whereas Neben, Bradley, and Hewitt (in preparation) demonstrate high dynamic range beam measurement using the constellation of ORBCOMM satellites. Here, we present in situ primary beam measurements of the MWA bow-tie antennas using the ORBCOMM constellation. We take advantage of both the high signal-to-noise ratio of ORBCOMM signals, and of our full cross-correlation measurements (rather than auto-correlations alone) to determine the beam.

In order to measure their primary beam profile $B_{\text{obs}}(r)$, we compare measurements with MWA antennas to simultaneous measurements with simple center-fed dipoles, whose beam pattern $B_{\text{dipole}}$ is known analytically. When there is a single extremely bright point source in the sky, such as an ORBCOMM satellite, we can compute the ratio of the visibilities of select baselines to obtain the ratio of the MWA antenna beam to the analytically known dipole antenna beam—thus determining the MWA antenna beam itself. To perform this analysis, two dipole antennas, one oriented along the $x$-polarization axis of the array and the other along the $y$-polarization axis, are added to the array and cross-correlated with all other MWA antennas.

20 We allow each model to have a separate weight to guard against potential calibration offsets between existing models.
Figure 13. Waterfall plots of phases on the 6 m E-W baseline. These show that our absolute calibration successfully matches the data (panel (a)) with a linear combination of global sky model and known point sources, including the Sun (panel (b)). Panel (c) shows the global sky model alone. The white areas are flagged out using $\chi^2$ criterion described in Figure 10. Each plot is stitched together using four independently measured and calibrated frequency bands, aligning local sidereal time. Thus the discontinuities between hours 4 and 12 are due to the Sun rising at different local sidereal times on different days of our observing expedition.

The rationale behind this technique is as follows. At an angular frequency $\omega$, the electric field from a sky signal at the position of a receiving antenna can be encoded in the Jones vector $S(\hat{k})$, where $\hat{k}$ is the position vector of the source (Collett 2005). With a primary beam matrix $B_j(\hat{k})$, the signal measured by the $j$th antenna at position $r_j$ is

$$s_j = \int e^{-i(k \cdot r_j + \omega t)} B_j(\hat{k}) S(\hat{k}) d\Omega.$$  \hfill (15)

When a single ORBCOMM satellite is above the horizon its signal strength is so dominant at its transmit frequency that $S(\hat{k})$ becomes well-approximated by a point source at the satellite’s location. The measured signal can then be written as:

$$s_j \approx e^{-i(k_s \cdot r_j + \omega t)} B_j(\hat{k}_s) S_s,$$  \hfill (16)

where $k_s$ is the wave vector of the satellite signal, and $S_s$ is the Jones vector encoding the satellite signal strength.

If we limit our attention to either $x$-polarization or $y$-polarization and approximate the off diagonal terms of $B(\hat{k})$ as zero, the visibility for two antennas can be written as

$$v_{jk} \approx S^2 B_j(\hat{k}_s)^* B_k(\hat{k}_s) e^{-ik_s \cdot (r_k - r_j)}.$$  \hfill (17)

If we take one visibility $v_{ij}$ formed by correlating a simple center-fed dipole with an MWA bow-tie antenna and another visibility $v_{kl}$ for the same baseline vector formed by...
Figure 14. Projected trajectories of 248 passes of ORBCOMM satellites over 40 hours. With these passes we obtain sufficiently dense sampling of the MWA antennas primary beam that we can robustly map its response, especially at high elevations where the response is strongest. With a map of the southern half of the primary beam, we can use the reflection symmetry of the antennas to infer the entire beam at the ORBCOMM transmission frequencies. Each curve is a satellite pass projected onto the x-y plane, and the different colors specify sets of data taken at different times.

correlating two MWA antennas, then their ratio is simply

$$\frac{|v_{ij}|}{|v_{kl}|} \approx \frac{|B_{\text{mwa}}(\hat{k}_s)|}{|B_{\text{dipole}}(\hat{k}_s)|},$$

(18)
because the satellite intensity $S$ and one MWA beam factor $B_{\text{mwa}}$ all cancel out, and the phase factor $e^{-i\hat{k}_s \cdot (r_j - r_i)}$ is removed due to taking absolute values of the visibilities. This means that when a single point source dominates the sky, the ratio of visibility amplitudes is simply the ratio of the antenna beams at the direction of the point source. Since we already know the beam $B_{\text{dipole}}$ of a center-fed dipole over a ground screen, we can directly infer the magnitude of MWA primary beam $|B_{\text{mwa}}(\hat{k}_s)|$.

In order to fully map out the MWA primary beam, we need to take data during many satellite passes until we have direction vectors that densely cover the entire sky. Satellite signals from 27 ORBCOMM satellites at 5 frequencies in the range of 137.2-137.8 MHz were identified. Their orbital elements are publicly available\(^{22}\), so we can calculate $\hat{k}_s(t)$ straightforwardly. With 40 hours of data taken at the frequencies of interest, we were able to obtain 248 satellite passes, shown in Figure 14.

We compared our measurements of the MWA primary beam using Equation (18) to numerical calculations using the FEKO electromagnetic modeling software package. Fixing an azimuth angle $\phi$, we can plot and compare the simulated and measured beam at different polar angles $\theta$ (the angle between the direction vector and zenith). Figure 15 shows how the beam changes with $\theta$ for four different $\phi$ values, where $\phi = 0$ correspond to North and increases clockwise. Our measurements of the MWA beams are seen to agree well with the numerical predictions for both polarizations. The small differences between the predicted and measured beams are larger than the statistical noise, implying that the main limitation is not noise but one or more of the above-mentioned approximations, or approximations in the electromagnetic antenna modeling.

3.2.3 Calibrating Array Orientation Using ORBCOMM Satellites and the Sun

The orientation of the array is very important, because the degeneracy removal process relies on the predicted measurement for each unique baseline, which in turn relies on precise knowledge of the baselines’ orientations. Although we measured the relative position of each antenna to millimeter level precision with a laser-ranging total station, we did not measure the absolute orientation of the array to better than the $\sim 1^\circ$ accuracy obtainable with a handheld compass. To improve upon this crude measurement, we make use of both the known positions of both ORBCOMM satellites and the Sun. As we show in Figure 16, the exceptional signal-to-noise in the ORBCOMM data allows us to fit for a small array rotation as a first order correction to a model based on our crude

\(^{22}\) We obtained the TLE files from CelesTrak, a company that archives TLEs of many civil satellites.
measurement. Our method for finding the true orientation of the array is as follows. For a given baseline during the peak few minutes of an ORBCOMM satellite pass at frequency $\nu$, we measure a phase $\phi(t)$. We also know the satellite’s position vector $k(t)$. However, we only have crude knowledge of baseline vector $d_0$ in units of wavelength, where vectors are in horizontal coordinates with $x, y, z$ that correspond to south, east and up. We can therefore only compute a crude prediction of the phase measurement

$$\phi_0(t) = 2\pi k(t) \cdot d_0.$$  \hfill (19)

We assume that the difference between the measurement $\phi(t)$ and our crude prediction $\phi_0(t)$ is due to a small angle rotation of the baseline vector $d_0$ around the axis $\theta = (\theta_x, \theta_y, \theta_z)$ by an angle $\theta = |\theta|$, ignoring a constant cable length delay.\(^{23}\) In the small $\theta$ regime, we have that

$$\phi(t) - \phi_0(t) = 2\pi k(t) \cdot (R(\theta) \cdot d_0 - d_0)$$

$$\approx 2\pi k(t) \cdot (\theta \times d_0)$$

$$= 2\pi (d_0 \times k(t)) \cdot \theta,$$  \hfill (20)

where $R(\theta)$ is the rotation matrix. Since we have a set of equations each representing a different time, the problem of finding $\theta$ can be reduced to that of finding a least squares fit. With 117 satellite passes, we obtained the following best fit for the array rotation around the vertical axis:

$$\theta^\text{sat}_z = 0.66^\circ \pm 0.0005^\circ_{\text{stat}} \pm 0.07^\circ_{\text{sys}}.$$

While this method is very precise for solving the main problem we were worried about—the direction of North ($\theta_z$) which we approximated in the field with a handheld compass—it is less useful for measuring rotations of the array in the other two directions. Our instrument’s absolute timing precision is only $\sim 0.5$ seconds, which makes it hard to distinguish rotations about the North-South axis from timing errors, as most ORBCOMM passes are East-West. This issue can of course be easily addressed in future experiments; for our experiment, we solve it using a more slowly moving bright point source: the Sun.

By using one day of solar data at 139.3 MHz, we obtained

$$(\theta_x, \theta_y, \theta_z) = (0.003^\circ, 0.003^\circ, 0.005^\circ),$$

$$\pm (0.01^\circ, 0.03^\circ, 0.004^\circ)_{\text{stat}}$$

$$\pm (0.04^\circ, 0.003^\circ, 0.005^\circ)_{\text{sys}}.$$

Although solar data is noisier, in part because the Sun is not as bright as the ORBCOMM satellites in a given channel, timing errors are no longer important. These results agree with and complement the satellite-based results and allow us to confidently pin down the orientation of the array and thus improve the quality of the calibration of all of our data. The excellent agreement between the independent measurements $\theta^\text{stat}_x \approx 0.66^\circ$ and $\theta^\text{stat}_z \approx 0.67^\circ$ provides encouraging validation of both the satellite and solar calibration techniques.

### 3.3 Systematics

As we discussed in section 3.1.4, most of our data are well-calibrated with $\chi^2/\text{DoF} < 1.2$, which means that any systematic effects should lie well below the level of the thermal noise. In this section we aim to identify all the systematic effects present in the system, and describe our efforts to quantify and, whenever possible, remove them. The systematics can be categorized into two groups:

(i) Signal-dependent systematics that grow as the signal becomes stronger, such as cross-talk, antenna position errors and antenna orientation errors.

(ii) Signal-independent systematics, such as radio frequency interference (RFI) from outside or inside the instrument.

\(^{23}\) Here it is important to use data before redundant baseline calibration to avoid phase degeneracy. We remove the phase delay from cables by allowing a constant offset that matches $\phi(t_M)$ with the crude prediction at time $t_M$ when the satellite has the strongest signal during the pass.
Below we find a strict upper bound of 0.15% for the signal-dependent component, as well as a signal-independent component which is easy to remove.

To quantify signal-dependent systematics, we again use ORBCOMM satellite data. Because the ORBCOMM signals are $10^3$ times brighter than astronomical signals, and we know that any signal-independent systematics must be weaker than the astronomical signals (otherwise they would have been blatantly apparent in the data), any signal-independent systematics must be negligible compared to the ORBCOMM signal. We therefore investigate how the discrepancies between calibrated visibilities and the models for each unique baseline depend on ORBCOMM signal strength. We define the average fitting error per baseline at a given time and frequency to be

$$\epsilon = \langle |v_{ij} - y_{i-j}g_i^*g_j| \rangle,$$

which is a combination of antenna noise and systematic errors. If we compute $\epsilon$ at different times with different signal strength and compute its signal dependence, we can derive an upper bound on the signal dependence of systematic errors. To do this, we take all data at the ORBCOMM satellite frequency over a day and compute $\epsilon$ after performing redundant calibration. We then bin the $\epsilon$-values according to the average signal strength $s = \langle |y_{i-j}| \rangle$, and obtain the results shown in Figure 17. The result is seen to be well fit by a constant noise floor plus a straight line $\epsilon \approx 0.0015s$. This slope implies that the combined effect of all signal-dependent systematic effects is at most 0.15%. This is merely an upper bound on the systematics, since it is possible that the increase in $\epsilon$ is mainly due not to systematics but to an increase in instrumental noise caused by an increase in the system temperature during the ORBCOMM passes.

There is one signal-dependent systematic that is not included in the above analysis: deviation from redundancy caused by imperfect positioning of antenna elements. This is because the data we used to derive the upper bound is always dominated by a single point source, the ORBCOMM satellite, and redundant calibration cannot detect any deviation of antenna position when the sky is dominated by a single point source. We have two ways to quantify the error in our data due to antenna position errors. Firstly, the laser-ranging measurements of antenna positions in the field indicate an average of 0.037 m deviation from perfect redundancy, which translates to about 2% average error in phase on each visibility. Since the deviations are in random directions, the variance of phase error in the unique baseline fits should be brought down by a factor equal to the number of redundant baselines, resulting in phase errors much less than 1% for most of the unique baselines. Secondly, although satellite calibration cannot detect position error in a given snapshot, over time the position errors would create very rapidly changing calibration parameters, which we do not observe in our data. Lastly, a formalism exists to treat errors in antenna placement as small perturbations when redundantly calibrating, although though we did not need to take advantage of this technique for the present paper.

We first identified a signal-independent systematic when we obtained consistent $\chi^2/\text{DoF} \sim 4$ for much of our data, which means that the fitting error was on average twice as large as the thermal noise in each visibility. This implies a systematic (or a combination of systematics) at the level of $10^{-6}$ pW/kHz, about 10% of the total astronomical signal. Given the above analysis, we can exclude the possibility of any signal-dependent explanations such as cross-talk between channels or antennas. While we are unable to offer any conclusive explanation of this systematic, it appears consistent with persistent near-field RFI, perhaps originating from our electronics. Fortunately, we found this additive signal to vary only very slowly over time, typically remaining roughly constant over 5-minute periods, which made it easy to remove. After calibrating the data with logistic, we average the fitting errors $\epsilon_{ij} = \langle v_{ij} - y_{i-j}g_i^*g_j \rangle$ over time and subtract them from the data before we run logistic

$$26$$ This was before we obtained a consistent $\chi^2/\text{DoF} \sim 1$ in Section 3.1.4 which occurred after we were able to remove the systematic described here.
again. We perform the averaging over 5 minute segments, corresponding to 112 independent time samples, and iterate the calibration-subtraction process three times. This corresponds to less than a 1% increase in the number of effective calibration parameters we fit for during logcal. Because many baselines probe the same unique baseline, the procedure described above exploits the redundancy of the array to robustly remove this slowly varying, signal-independent systematic, leaving us with $\chi^2/\text{DoF} \sim 1$.

4 SUMMARY AND OUTLOOK

We have described the MITEoR experiment, a pathfinder “omniscope” radio interferometer with 64 dual-polarization antennas in a highly redundant configuration. We have demonstrated a real-time precision calibration pipeline with automatic data quality monitoring that uses $\chi^2/\text{DoF}$ as a data quality metric to ensure that redundant baselines are truly seeing the same sky. We have also implemented various instrumental calibration techniques that utilize the ORBCOMM constellation of satellites to measure the primary beam and precise orientation of the array. Our success bodes well for future attempts to perform such calibration in real-time instead of in post-processing, and thus clears the way for FFT correlation that will make interferometers with $\gtrsim 10^3$ antennas cost-efficient by reducing the computational cost of correlating $N$ antennas from an $N^2$ scaling to an $N \log N$ scaling. It also suggests that the extreme calibration precision required to reap the full potential of 21 cm cosmology is within reach.

The various calibration techniques that MITEoR successfully demonstrates are now being incorporated into the much more ambitious HERA project\cite{Pober+2014}, a broad-based collaboration among US radio astronomers from the PAPER, MWA, and MITEoR experiments. Our results are also pertinent to the design of the SKA low-frequency aperture array\cite{HERA}. HERA plans to deploy around 331 14-meter dishes in a close-packed hexagonal array in South Africa, giving a collecting area of more than 0.05 km$^2$, virtually guaranteeing not only a solid detection of the elusive cosmological 21 signal but also interesting new clues about our cosmos.

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**APPENDIX A: PHASE DEGENERACY IN REDUNDANT CALIBRATION**

Both of our redundant baseline calibration algorithms, logcal and lincal (see Section 3.1.3), have the same set of phase degeneracies that require additional absolute calibration that must incorporate knowledge of the sky. When calibrating a given unique baseline, the quantity that logcal minimizes is

$$
\sum_{jk} (|\theta_{j,k} - \phi_j + \phi_k| - \arg(v_{jk}))^2,
$$

(A1)

where we define $\theta_{j,k} \equiv \arg(y_{j,k}), \phi_j \equiv \arg(g_j)$. Similarly, lincal minimizes

$$
\sum_{jk} (|y_{j,k} g_j g_k^* - v_{jk}|^2 = \sum_{jk} |(y_{j,k} g_j g_k^* \exp[i(\theta_{j,k} - \phi_j + \phi_k)] - v_{jk}|^2).
$$

(A2)

Unfortunately, for all values of $\theta_{j,k}$ and $\phi_k$, one can add any linear field $\vec{\Phi} \cdot \vec{r}_j + \psi$ to the $\phi_j$ across the entire array while subtracting $\vec{\Phi} \cdot \vec{d}_j$ from the $\theta_{j,k}$ without changing the minimized quantities:

$$
\begin{align*}
\theta'_{j,k} - \phi'_{j,k} & \equiv (\theta_{j,k} - \vec{\Phi} \cdot \vec{r}_j + \psi) - (\phi_j + \vec{\Phi} \cdot \vec{d}_j + \psi) \\
& \equiv \theta_{j,k} - \phi_j - \phi_k.
\end{align*}
$$

(A3)

Here $\vec{r}_j$ is the position vector of antenna $j$ and $\vec{d}_{j,k} \equiv \vec{r}_k - \vec{r}_j$ is the baseline vector for the unique baseline with best-fit visibility $y_{j,k}$. Thus, the quantities in expressions [A1] and [A2] that the calibrations minimize are degenerate with changes to the linear phase field $\vec{\Phi}$ and the scalar $\psi$. This means that there are, in general, 4 degenerate phase parameters that need absolute calibration: one overall phase $\psi$ and three related to the three degrees of freedom of the linear function $\vec{\Phi}$ (which reduces to two for a planar array).

In an ideal instrument, the measured visibilities for a given unique baseline would be

$$
y_{i,j} = \int_{k_x, k_y} e^{ik \cdot \vec{d}_{i,j}} S_B(k_x, k_y) dk_x dk_y,
$$

(A4)

where $k = (k_x, k_y, k_z)$ is the wave vector of incoming radiation and $S_B(k_x, k_y)$ is the product of the incoming signal intensity and the primary beam in the direction $\vec{k}$ normalized by $kk_z$ (which comes from the Jacobian of the coordinate transformation from spherical coordinates; see [Tegmark & Zaldarriaga 2000]). When the array is coplanar, we can take an inverse Fourier transform of $y_{i,j}$ and obtain an image of $S_B(k_x, k_y)$. Above we saw that the best fit $y_{i,j}$ computed by logcal and lincal is multiplied by an unknown linearly varying phase $\vec{\Phi} \cdot d_{i,j}$. Since multiplication in $\mathbf{w}$ space is a convolution in image space, this means that the image generated using these $y_{i,j}$ is the true image convolved with a Dirac delta function centered at $\vec{\Phi}$, which corresponds to a simple shift by the unknown vector $\vec{\Phi}$ in the $S_B(k_x, k_y)$ image space.

To calibrate these last few overall phase factors, one can either make sure that bright radio sources line up properly in the image, or match phases between measured visibilities and predicted visibilities, as we described in Section 3.2.1. However, there may be another complementary way to remove this phase degeneracy without any reference to
the sky. We know that physically the true image \( S_{\Phi}(k_x, k_y) \) is only non-zero within the disk \( |k_x^2 + k_y^2|^{1/2} \leq k \) centered around the origin, and a shift caused by \( \Phi \) would move this circle off center. This suggests that we should be able to reverse engineer \( \Phi \) by looking at how much the image circle has been shifted, without knowing what \( S_{\Phi}(k_x, k_y) \) is. Figure A2 demonstrates how the image is shifted by \( \Phi \) using simulated data.

Unfortunately, this simple approach to identifying and removing the effect of \( \Phi \) suffers from a few complications. By far the most important one is the requirement of very short baselines. In the example in Figure A1 the shortest separation between antennas is 0.21\( \lambda \), and it is easy to show that the sky disk is only clearly demarcated when the shortest separation is less than 0.5\( \lambda \). This sets a limit on the physical size of each element, which makes achieving a given collecting area proportionately more difficult. As Figure A2 shows, the deployed configuration of MITTEoR cannot be used to reverse engineer the degeneracy vector \( \Phi \) without knowledge of the true sky.

**Figure A1.** Illustration of phase degeneracies shifting the sky image where the sky disk is demarcated. The linear phase degeneracy, which takes the form \( \Phi \cdot r_j \) in each antenna for any \( \Phi \), corresponds to a shift of the reconstructed image. These simulated images demonstrate shifts of fiducial sky image at 160 MHz caused by four different \( \Phi \), where the fiducial array’s shortest baseline is 0.2 m. Panel (a) shows the image obtained from visibilities with \( \Phi = (0, 0) \), and the sky image is centered at 0. In the other three panels, the sky image is shifted by amount \( \Phi \). Even if one has no knowledge of what the true sky is, it is still possible to determine \( \Phi \) from where the sky image is centered.

**Figure A2.** Illustration of phase degeneracies shifting the sky image where the sky disk is not demarcated. With any practical array configuration, including that of MITTEoR, distinguishing image shifts caused by the \( \Phi \)-degeneracy becomes significantly more difficult. These images demonstrate shifts of fiducial sky image at 160 MHz just as in Figure A1 but with MITTEoR’s compact configuration where the shortest baseline is 1.5 m. In the left panel, the image is obtained from visibilities with \( \Phi = (0, 0) \), and in the right panel the sky image is shifted by amount \( \Phi = (0, 0.3k) \).

**APPENDIX B: A HIERARCHICAL REDUNDANT CALIBRATION SCHEME WITH \( O(N) \) SCALING**

One of the major advantages of an omniscope is its \( N \log N \) cost scaling where \( N \) is the number of antennas. However, existing calibration techniques, including the ones presented in this paper, require all of the visibilities to compute the calibration parameters. Since the cost for computing the visibilities alone scales as \( N^2 \), this is a lower bound to the computational cost of existing calibration schemes regardless of the actual algorithm. While current instruments with less than 10\(^4\) elements can afford full \( N^2 \) cross-correlation, such computation will be extremely demanding for a future omniscope with 10\(^4\) or more elements. Thus, to take advantage of the \( N \log N \) scaling of an omniscope with large \( N \), it is necessary to have a calibration method whose cost scaling is less than \( N \log N \). In this section, we describe a such a method using a hierarchical approach, and show that its computational cost scales only linearly with the number of antennas.

Figure B1 illustrates the hierarchical calibration method for an example with a 256 antennas in a 16 \times 16 regular grid, viewed as a 2-level hierarchy of 4 \times 4 grids. More generally, consider an \( n \)-level hierarchy with \( m \) sub-arrays at each level, containing a total of \( N = m^n \) antennas; the example in Figure B1 corresponds to \( m = 16 \), \( n = 2 \) and \( N = 256 \). Let \( B_m \) denote the computational cost of calibrating a basic \( m \)-antenna array\(^{30}\). Let \( C_n \) denote the computational cost of calibrating the entire \( n \)-level hierarchy

\[ C_n = m C_{n-1} + B_m \]

\(^{29}\) This is the 2D imaging counterpart of the well-known fact that, in signal processing, one must sample with a time interval shorter than 0.5\( \nu^{-1} \) to avoid aliasing in the spectrum of maximum frequency \( \nu \).

\(^{30}\) It is easy to see that for a regular grid of \( N \) antennas, \( N \) need not an exact power of \( m \) to obtain the scaling that we will derive.

\(^{31}\) \( B_m \) includes the cost to compute cross-correlations between the \( m \) antennas, as well as both relative and absolute calibra-
visibility is computed between each pair of sub-arrays, rather than 256 visibilities in a full correlation scheme. Because of the reduced number of cross-correlations computed, we expect the quality of calibration parameters to be worse than that in the full correlation case. Since both the precision of calibration parameters and the computational cost depend on $m$, one can tune the value of $m$ to achieve an optimal balance between precision and computational cost.

**APPENDIX C: FAST ALGORITHM TO SIMULATE VISIBILITIES USING GLOBAL SKY MODEL**

For both traditional self-calibration and the absolute calibration described in this paper, it is crucial to have accurate predictions for the visibilities. This requires simulation of both the contributions of bright point sources and diffuse emission, which can be added together due to the linearity of visibilities. While it is computationally easy to compute the contributions of point sources of known flux, it is much harder to compute visibilities from diffuse emission such as that modeled by the global sky model (GSM, de Oliveira-Costa et al. 2008). Dominated by Galactic synchrotron radiation, this diffuse emission is especially important for the low frequencies and angular resolutions typical of current 21 cm experiments.

In general, we want to compute visibilities

$$ y_u(f,t) = \int s(\hat{k}, f, t) B(\hat{k}, f) e^{i k_d u} d\Omega_k, $$

where $s(\hat{k}, f, t)$ is the magnitude squared of the global sky model at time $t$ in horizontal coordinates, and $B(\hat{k}, f)$ the magnitude squared of the primary beam at a given frequency. Performing the integral by summing over all $n_{\text{pix}}$ pixels in the GSM takes $O(n_{\text{pix}} n_f n_t)$ computational steps, where $n_{\text{pix}}$ is the number of unique baselines one simulates, $n_f$ is the number of frequency bins, and $n_t$ is the number of visibilities one simulates for one sidereal day.

The faster algorithm we describe here takes only $O(n_{\text{pix}} n_f n_t)$ steps, by taking advantage of the smoothness of the primary beam as well as the diurnal periodicity in Earth’s rotation. It applies only to drift-scanning instruments, so $B(\hat{k}, f, t) = B(\hat{k}, f)$ in horizontal coordinates, and is similar in spirit to the ideas proposed by Shaw et al. (2014).

The key idea is to decompose Equation (C1) as follows:

$$ y_u(f,t) = \sum_{l,m} a_{lm}^f B_{lm}^f e^{i m \phi(t)}, $$

where each $a_{lm}^f$ is a spherical harmonic component of the GSM at a given frequency, and each $B_{lm}^f$ is a spherical harmonic component of $B(\hat{k}, f) e^{i k_d u}$, both in equatorial coordinates. In this appendix, we describe precisely how to perform this decomposition and why it decreases the computational cost of calculating visibilities from the GSM.

### C1 Spherical Harmonic Transform of the GSM

The GSM of [de Oliveira-Costa et al. 2008](http://example.com) is composed of three HEALPix maps of size $n_{\text{side}}$ describing different frequency-independent sky principal components $s^f(\hat{k})$ and...
the relative weights of each component $w^c(f)$ that encode the frequency dependence. We can decompose the spatial dependence into spherical harmonics,

$$a_{\ell m}^c = \int Y_{\ell m}^*(\hat{k}) s^c(\hat{k}) d\Omega_k$$  \hfill (C3)

in $O(n_{\text{f}}^3)$ steps, due to the advantage of HEALPix format (Górski et al. 2005). The frequency dependence of the spherical harmonic coefficients of the sky is given by

$$a_{\ell m}^f = \sum_c a_{\ell m}^c w^c(f),$$ \hfill (C4)

and the total complexity of computing the coefficients $a_{\ell m}^f$ is $O(n_{\text{pix}}^2) + O(n_{\text{f}})$.

### C2 Spherical Harmonic Transform of the Beam and Phase Factors

Next, we would like to compute the spherical harmonics components of $B(\hat{k}, f) e^{i k \cdot d_u}$:

$$B_{\ell m}^{sf} = \int Y_{\ell m}^*(\hat{k}) B(\hat{k}, f) e^{i k \cdot d_u} d\Omega_k.$$  \hfill (C5)

Substituting the spherical harmonic decompositions of $B(\hat{k}, f)$ and $e^{i k \cdot d_u}$ gives

$$B_{\ell m}^{sf} = \int Y_{\ell m}^*(\hat{k}) \sum_{\ell', m'} B_{\ell', m'}^f Y_{\ell', m'}(\hat{k})$$

$$\times \sum_{\ell'' m''} 4\pi \ell'' j_{\ell''} \left( \frac{2\pi f}{c} d_u \right) Y_{\ell'' m''}(\hat{d}_u) Y_{\ell' m'}(\hat{k}) d\Omega_k$$

$$= \sum_{\ell, m} 4\pi \ell j_{\ell} \left( \frac{2\pi f}{c} d_u \right) B_{\ell m}^f Y_{\ell m}^*(\hat{d}_u)$$

$$\times \sum_{\ell, m} 4\pi \ell j_{\ell} \left( \frac{2\pi f}{c} d_u \right) B_{\ell m}^f Y_{\ell m}^*(\hat{k}) d\Omega_k$$

$$\times \sum_{\ell, m} 4\pi \ell j_{\ell} \left( \frac{2\pi f}{c} d_u \right) B_{\ell m}^f Y_{\ell m}^*(\hat{d}_u)$$

$$\times \sum_{\ell, m} 4\pi \ell j_{\ell} \left( \frac{2\pi f}{c} d_u \right) B_{\ell m}^f Y_{\ell m}^*(\hat{k}) d\Omega_k$$

$$\times \left( \frac{(2\ell + 1)(2\ell' + 1)(2\ell'' + 1)}{4\pi} \right)$$

$$\times (-1)^m \left( \begin{array}{ccc} \ell & \ell' & \ell'' \\ 0 & 0 & 0 \\ -m' & m' & m'' \end{array} \right),$$ \hfill (C6)

where $j_{\ell}(x)$ is the spherical Bessel function, $\ell' m'$ represent quantum numbers when expanding the primary beam, $\ell'' m''$ represent quantum numbers when expanding $e^{i k \cdot d_u}$, and the Wigner-3j symbols are results of integrating the product of three spherical harmonics. Because the Wigner-3j symbols vanish unless $\ell = \ell' \leq \ell'' \leq \ell + \ell'$ and $-m + m' + m'' = 0$, the above sum simplifies to

$$B_{\ell m}^{sf} = \sum_{\ell', m'} 4\pi \ell j_{\ell} \left( \frac{2\pi f}{c} d_u \right) B_{\ell m}^f Y_{\ell m}^*(\hat{d}_u)$$

$$\times \sqrt{\frac{(2\ell + 1)(2\ell' + 1)(2\ell'' + 1)}{4\pi}}$$

$$\times (-1)^m \left( \begin{array}{ccc} \ell & \ell' & \ell'' \\ 0 & 0 & 0 \\ -m' & m' & m'' \end{array} \right),$$ \hfill (C7)

where $m'' = m - m'$. Note that $\ell', m'$ and $\ell''$ in this sum are all limited to the range of $\ell$-values where the spherical harmonics components for the primary beam are non-zero, so the complexity for this triple sum is $n_{\text{pix}}^3$, where $n_{\text{pix}}$ is the number of non-zero spherical harmonics components for the primary beam. Since the cost for each $B_{\ell m}^{sf}$ is $n_{\text{pix}}^2$, and there are $n_{\text{f}} n_{\text{pix}}$ of them, the computational complexity of calculating all $B_{\ell m}^{sf}$ coefficients scales like $O(n_{\text{f}} n_{\text{pix}} n_{\text{pix}}^2)$.

### C3 Computing Visibilities

By performing a coordinate transformation on Equation [C1] from horizontal coordinates (corresponding to the local Horizon at the observing site) to equatorial coordinates, the time dependence of $s(k)$ is transferred to $B(k)$ and $d_u$. We can now calculate $y_u(f, t)$ by computing

$$y_u(f, t) = \int s(k) B^f(k) e^{i k \cdot d_u(t)} d\Omega_k$$

$$= \sum_{\ell m} a_{\ell m}^f B_{\ell m}^{uf}.$$ \hfill (C8)

Since the time dependence of $B_{\ell m}^{uf}$ is a constant rotation along the azimuthal direction, we can write the above as

$$y_u(f, t) = \sum_{\ell m} a_{\ell m}^f B_{\ell m}^{uf} e^{i m\phi(t)} = \sum_{m} c_{m}^f e^{i m\phi(t)},$$ \hfill (C9)

where we have defined

$$c_{m}^f \equiv \sum_{\ell} a_{\ell m}^f B_{\ell m}^{uf},$$ \hfill (C10)

which can be computed in $O(n_{\text{f}} n_{\text{pix}}^2)$ steps. Given $c_{m}^f$, it is clear that we can evaluate Equation [C9] using a fast Fourier Transform (FFT), whose cost is

$$O(n_{\text{f}} n_{\text{pix}} \log(n_{\text{pix}})).$$ \hfill (C11)

Note that this FFT in Equation [C9] has no $n_{\text{pix}}$ dependence, because we always need to zero-pad $c_{m}$ to length $n_t$ before the FFT. In summary, the total complexity of all of the above steps is

$$O\left(n_{\text{pix}}^3\right) + O(n_{\text{f}}) + O\left(n_{\text{f}} n_{\text{pix}} n_{\text{pix}}^2\right)$$

$$+ O\left(n_{\text{f}} n_{\text{pix}}\right) + O\left(n_{\text{f}} n_{\text{pix}} \log(n_{\text{pix}})\right)$$

$$\approx O\left(n_{\text{f}} n_{\text{pix}} n_{\text{pix}}^2\right).$$ \hfill (C12)

This does not scale with $n_t$, unlike the naive integration’s $O(n_{\text{f}} n_{\text{pix}} n_{\text{pix}} n_{\text{pix}})$. Thus with a spatially smooth beam whose $n_{\text{pix}} \ll n_{\text{pix}}^2$, the algorithm described here is much faster than the naive numerical integration approach described at the beginning of this Appendix.